Companies Specials (8)
IP Telephones (36)
Wi-Fi Telephones (6)
Voip Fax (2)
Skype Telephones (5)
Smartphones with Voip (5)
Mobile Telephones (1)
Analogue Telephone Adapter (9)
Voip Router (10)
Voip Gateways (7)
IP-PBX (2)
MeshVPN (3)
Asterisk
DECT with VoIP (6)
WiFi Radio (1)
ISDN cards (1)
Headsets (8)
International numbers (1)
Voip accessories (1)
Hosted PBX Weblink (1)

Companies Specials

Siemens S450IP with GN Netcom 9120 headset
Modèle S4509120
Voicetech Sweden
Siemens S450IP telephone with a GN Netcom 9120 headset bundle.
Svenska kronor2000.00

SNOM 360 Package 360GN
Modèle 360GN
Snom
SNOM 360 + GN Netcom 9120 FlexBoom + EHS Wireless Headset Adapter Bundle   With this bundle you can take incoming call directly on the ear, even if you are not situated at your working station.   Bundle consisting of: ---------- SNOM 360 IP Telefon   GN Netcom 9120 FlexBoom   SNOM Wireless Headset Adapter   ----------
~~Svenska kronor6800.00~~ Svenska kronor4320.00

SNOM 360 Package Netcom 9350
Modèle 360GN9350
SNOM 360 + GN Netcom 9350 + EHS Wireless Headset Adapter Bundle   With this bundle you can take incoming call directly on the ear, even if you are not situated at your working station.   Bundle consisting of: ---------- SNOM 360 IP Telefon   GN Netcom 9350 SNOM Wireless Headset Adapter   ----------

SNOM 360

Features

Tiltable graphical display (128x64 pixels)
47 keys, 13 LEDs
12 programmable function keys
Speakerphone
Dual Ethernet connection
Power over Ethernet
Headset connection
Additional keypads with 42 programmable function keys
SIP RFC3261
Security: SIPS/SRTP STUN, ENUM, NAT, ICE
Compression: G.723.1 and others
National Language Support
Asian Language Support

GN NETCOM 9350 NC Funk Mono 9356-607-101
GN 9350 is the ultimate example of wireless office headset functionality. For use with both traditional desk telephones and IP telephony, the dual connection GN 9350 always meets your needs.GN 9350 makes everything sound better. State-of-the-art Digital Signal Processing (DSP) technology removes impurities from the incoming signal, enriches sound quality, and maintains a safe consistent volume level. The GN 9350?s sleek boom arm with noise-cancelling microphone means your voice is transmitted clearly, even in noisy environments.When used with IP telephony applications, GN 9350 is the first wireless headset to provide wideband audio, with an astounding 6.8 kHz range ? twice that of conventional telephony.With a wireless range of up to 100 metres, you can roam your office to your heart?s content, whilst an optional nine-hour, user-replaceable battery means you can talk around the clock.Features: * DSP and IntelliTone? for superb sound quality and greater hearing protection   * Future-proof investment thanks to built-in USB interface for PC-based IP   telephone   * LCD display for easy set-up of personal sound preferences   * Noise-cancelling microphone for reduced background noise   * Around-the-clock talk time with optional second battery   * Light-weight design and multiple wearing styles -neckband included   * Multi-unit conferencing capability

SNOM Wireless Headset Adapter

snom wireless headset adaptor for GN Netcom and Plantronics EHS Headsets. Suitable for 320, 360, 370 snom phones.

Snom Wireless Headset Adapter for Snom 320/Snom 360/Snom 370
~~Svenska kronor6000.00~~ Svenska kronor4560.00

SNOM 370 Package 370GN
Modèle 370GN
Snom
SNOM 370 + GN Netcom 9350 + EHS Wireless Headset Adapter Bundle With this EHS-bundle you can take incoming call directly on the ear , even if you are not situated at your working station. (EHS = Electronic Hook Switch) Bundle consisting of: ---------- SNOM 370 IP Telefon   GN Netcom 9350 SNOM Wireless Headset Adapter   ----------
~~Svenska kronor6000.00~~ Svenska kronor4800.00

SNOM 370 Package Midiboom
Modèle 370MID
Snom
SNOM 370 + GN Netcom 9120 MidiBoom + EHS Wireless Headset Adapter Bundle   With this EHS-bundle you can take incoming call directly on the ear , even if you are not situated at your working station. (EHS = Electronic Hook Switch) Bundle consisting of: ---------- SNOM 370 IP Telefon   GN Netcom 9120 MidiBoom SNOM Wireless Headset Adapter   ----------
~~Svenska kronor5600.00~~ Svenska kronor4320.00

IP Telephones

Aastra 480i IP telephone
Modèle 480I
Aastra
Large display — Eight-line backlit screen with six context sensitive softkeys provides more information and flexibility in call handling Enhanced Call Management — Large storage for personal directory, callers log, and redial list Tight Integration — Support for Spherical MGCP and multiple SIP IP telephony systems including BroadWorks®, Nortel, Sylantro and Asterisk Multiple Line/Call Appearance Lights — Support up to 9 simultaneous calls. Juggle multiple calls easily with a single press of a button Shared Call Appearances — Shared calls allow you to place a call on hold at one set and pick it up easily at another set Remarkable Audio — Quality speakerphone with excellent voice clarity and delivery Protect Your Investments — Firmware upgrades can be downloaded and installed in the field as standards develop and protocols evolve Less Wiring — Built in dual 10/100 switched Ethernet ports let you share a connection with your computer. Inline power support eliminates power adapters
~~**Svenska kronor1680.00**~~ **Svenska kronor1520.00**

Aastra 9112i IP telefon
Modèle 9112I
Aastra
Enhanced Call Management — Large storage for personal directory, callers log, and redial list Tight Integration — Support for multiple IP telephony systems including BroadWorks®, Nortel, Sylantro, and Asterisk SIP Remarkable Audio — Quality speakerphone with excellent voice delivery Protect Your Investments — Firmware upgrades can be downloaded and installed in the field as standards develop and protocols evolve Feature Keys 4 navigational keys 2 programmable keys 11 predefined hard keys including Callers log, Conference, Call Transfer, Redial, Options, Directory, Save, Delete, Speaker/Headset, Mute Networking 10/100 Mbps Ethernet port Manual or Dynamic Host Configuration Protocol (DHCP) IP address setup Time and date synchronization using SNTP Built-in HTTP server for web administration and maintenance Server provisioned user configuration files Protocols / Codecs IETF SIP (RFC3261) G.711 μ-law / A-law G.729
Svenska kronor880.00

Allnet 7960 IP telephone
Modèle ALL7960
Allnet
- Integrated Phonebook with 200 numbers storageroom- 10 programmable Hotkeys - Handsfree function- Conferencecall function - 10/100MB Switch connection- 2-part display - Number recognition- STUN en UPnP - Quality of Service - Configurable through webbrowser - Digital signal processor DSP - Noice reduction
Svenska kronor800.00

Elmeg IP290 IP Telephone
Modèle IP290
Elmeg
The Elmeg IP290 is identical in construction with Snom´s 190 phone. Features:2 x 24 Digit Display3 Softkeys and 5 programmable function keysHeadset enabledRFC 326-compatible SIP-StackDual EthernetMultiple language supportPhonebook for 100 numbersCall waitingCall forwardingCalling via shortcodeCall logDisplays waiting calls Datasheet Elmeg IP290
Svenska kronor1080.00

Grandstream Budgetone 200 IP telephone
Modèle BT200
Granstream Networks
The Grandstream BT200 offers extensive comfort- functions at a small price. In addition to a full- duplex speakerphone the BT200 provides a 2,5mm headset jack. Furthermore it supports common audio codecs and standard functions like Transfer, Hold and 3-way conferencing. The BT200 also features easy configuration via web interface. Through the second Ethernet port (100Base- T) you can also connect your PC directly to the BT200. Main features - full- duplex speakerphone - 2,5mm headset jack - 2 Ethernet ports Technical specifications - Standards: IP, TCP, UDP, RTP/RTCP, HTTP, ARP/RARP, DNS, DHCP,   NTP, TFTP, PPPoE - Connection Port: 2x RJ45 (100Base- T, switched or routed)   Ethernet port, 2,5mm headset jack - Protocol: SIP - Audio CODEC: G.711, G.729A/B, G.723.1, G.726, G.728, G.722 - Management: HTTP/HTTPS Web Interface, via keypad - Fax support: no - Dimensions: 180 mm x 220 mm x 65mm (WxDxH), 0,9 kg
Svenska kronor720.00

Grandstream GXP 2000 IP Telephone
Modèle GXP2000
Granstream Networks
Grandstream GXP-2000 is a next generation enterprise IP telephone based on open industry standards. Built on innovative technologies, GXP-2000 features market leading superb audio quality, rich functionalities, and excellent manageability at affordable prices. Main Features: Support SIP (RFC3261), TCP/IP/UDP, RTP/RTCP, HTTP, ARP, ICMP, DNS (A record and SRV), DHCP, PPPoE, TFTP, NTP Support NAT traversal (STUN, etc), server fail-over, SIP presence (SIMPLE), and more Multiline support of up to 11 lines indicators (expandable to a few dozen more through expansion key-module) Graphical LCD to display up to 8 lines and 22 characters per line Dual 10/100Mbps Ethernet ports Headset jack Support Caller ID display or block, per call or permanent Call waiting, Hold, Mute, Transfer (blind or attended), Forward, and more Multi-party conferencing Integrated Power-over-Ether (802.3af and Cisco style) And many more enterprise grade features   Asterisk Configuration GXP-2000 Manual GXP-2000
Svenska kronor960.00

Grandstream GXP-2010 IP telephone
Modèle GXP2010
Granstream Networks
In stock! The new GXP2010 Key System IP phone has a new sleek design and delivers excellent call quality and enterprise grade feature set that includes advanced XML capabilities, multi-party conferencing, multi-language support, presence and BLF (busy lamp field), security protection, automated provisioning, and broad compatibility with leading SIP platforms. The GXP2010 offers 4 lines, 18 programmable keys, 3 dynamic context-sensitive XML soft keys, dual switched 10M/100Mbps auto-sensing Ethernet ports with integrated PoE, and a large high-resolution backlit LCD display. As with all models in the GXP Series, this model supports comprehensive voice codecs and SRTP for privacy protection. The GXP2010 was designed to be the cost-effective choice for the busy call center or small business that wants the feature functionality of a key-system. The 18 programmable speed-dial keys enable one-button access to office personnel and the 11 dedicated keys create one-button access to indispensable telephony features including hold, conference, mute, do-no-disturb, intercom, transfer, speaker and voicemail. This phone is ideal for enhancing office productivity and improving efficiency. Technical Features & Benefits Support SIP (RFC3261), TCP/IP/UDP, RTP/RTCP, HTTP, ARP, ICMP, DNS (A record and SRV), DHCP(both client and server), PPPoE, TFTP, NTP Dual switched 10/100Mbps Ethernet ports w/integrated Power-over-Ethernet (802.3af) Support NAT traversal (STUN, etc), server fail-over, SIP presence (SIMPLE), and more Backlit 240x120 high resolution graphic LCD with multi-level gray scales

Call Features & Benefits 4 lines indicators with individual SIP account profiles 18 programmable hard keys and 3 XML programmable context-sensitive soft keys Multi-line support of up to 22 call appearance lines with dual-color LED indicators Support Caller ID display or block, per call or permanent Call waiting, hold, mute, transfer (blind or attended), forward, and more Multi-party conferencing (up to 4-way) and shared line appearances Expandable through expansion key-modules (pending) And many more enterprise grade features Handset Features & Benefits Full duplex speakerphone w/advanced audio echo cancellation Headset jack (2.5 mm and RJ11) for broad headset compatibility Secure and automated provisioning for mass deployment SRTP and TLS (pending) for privacy protection And many more enterprise grade features Feature Specifications SIP Compliant and Protocols: SIP, TCP/IP/UDP, RTP/RTCP, HTTP, ARP, ICMP, DNS (A record and SRV), DHCP(both client and server), PPPoE, TFTP, NTP, etc.Networking Interfaces: Dual 10/100mbps Ethernet ports (switched or routed).Voice Codecs: G.711 (a/μ-law), G.723.1, G.729A/B, G.726, GSM, iLBC, and G.722 (wideband).Superb Audio Quality: Advanced Digital Signal Processing (DSP), Silence suppression, VAD, CNG, AGC high fidelity wideband audio (G.722).Custom Ringtone: Software Convert most music files to a Grandstream ringtone. Click here to download.Advanced Functionality: Multi-line support, multi-party conferencing (4-way), multi-language support (MLS), headset enabled, expandable, intercom, AES encryption, etc.
Svenska kronor1600.00

Grandstream GXV-3000
Modèle GXV3000
Granstream Networks
Grandstream GXV-3000 is a next generation advanced IP video phone based on SIP and H.264 standard. Built upon Grandstream's innovative technology, GXV-3000 features superb audio and video quality, rich functionality, ease of use, stylish exterior design, and highly attractive price. Main Features: - Support SIP 2.0, TCP/IP/UDP, RTP/RTCP, HTTP/HTTPS, ARP/RARP, ICMP, DNS, DHCP (both client and server), PPPoE, TFTP, NTP, Telnet, TLS (pending), etc.- Support dual 10M/100M auto-sensing Ethernet ports configurable to operate under either switch or router mode.- Powerful video DSP with advanced adaptive jitter control and packet loss concealment technology to ensure superb audio and video quality.- Support advanced H.264 base line real-time video codec (at CIF or QVGA resolution and up to 30 frames/second)- Support popular voice features including 3 line indicators, full-duplex hands free speaker phone, 3-way conference, etc- Support advanced video features including:- High quality 5.6-inch TFT color LCD panel 2 dimensionallyrotable and camera tiltable to allow virtually all viewing angles- Advanced VGA resolution CMOS camera sensor- Anti-flickering- Auto focus and auto exposure- Zoom- PIP (Picture-in-Picture)- Audio mute and camera block (for privacy)- Still picture capture/store/send (VGA Resolution) - Support 2 USB (2.0) host ports, 1 audio and 1 video output jack and headset jack- Support device configuration via LCD, Web browser or central secure configuration file   Datasheet GXV 3000 Manual GXV 3000 Quick Install Guide GXV 3000
Svenska kronor2240.00

Granstream GXP-2020 IP telephone
Modèle GXP2020
Granstream Networks
 GXP-2020 6-line Enterprise IP Phone The GXP2020 Enterprise SIP phone sports a new stylish design and addresses the need for an elegant IP handset solution for the executive office at a highly competitive price. The GXP2020 provides excellent voice clarity, a comprehensive set of advanced call features, multi-language support, security protection, automated provisioning, and broad compatibility with leading SIP platforms. The GXP2020 features 6 lines, 7 programmable keys, 4 dynamic context-sensitive soft keys, dual switched 10M/100Mbps auto-sensing Ethernet ports with integrated PoE, a backlit 320x160 high resolution graphic LCD with multi-level grey scales, SRTP and TLS (pending) for privacy protection, as well as secure and automated provisioning for mass deployment. The GXP2020 is the second phone in the GXP Enterprise SIP Phone portfolio. It is ideal for both the executive office and advanced enterprise users. Features & Benefits Technical Features & Benefits Support SIP (RFC3261), TCP/IP/UDP, RTP/RTCP, HTTP, ARP, ICMP, DNS (A record and SRV), DHCP(both client and server), PPPoE, TFTP, NTP Dual switched 10/100Mbps Ethernet ports w/integrated Power-over-Ethernet (802.3af) Support NAT traversal (STUN, etc), server fail-over, SIP presence (SIMPLE), and more Backlit 320x160 high resolution graphic LCD with multi-level grey scales Call Features & Benefits 6 lines indicators with individual SIP account profiles 7 programmable hard keys and 4 XML programmable context-sensitive soft keys Multi-line support of up to 13 call appearance lines with dual-color LED indicators Expandable to additional 112 lines through expansion key-modules Support Caller ID display or block, per call or permanent Call waiting, hold, mute, transfer (blind or attended),forward, and more Multi-party conferencing (up to 5-way) And many more enterprise grade features Handset Features & Benefits Full duplex speakerphone w/advanced audio echo cancellation Headset jack (2.5 mm and RJ22) Secure and automated provisioning for mass deployment SRTP and TLS (pending) fro privacy protection And many more enterprise grade features Feature Specifications SIP Compliant and Protocols SIP, TCP/IP/UDP, RTP/RTCP, HTTP, ARP, ICMP, DNS (A record and SRV), DHCP(both client and server), PPPoE, TFTP, NTP, etc. Networking Interfaces Dual 10/100mbps Ethernet ports (switched or routed). Superb Audio Quality Advanced Digital Signal Processing (DSP), Silence suppression, VAD, CNG, AGC high fidelity wideband audio (G.722). Custom Ringtone Software Convert most music files to a Grandstream ringtone. Click here to download. Advanced Functionality Multi-line support, multi-party conferencing (5-way), multi-language support (MLS), headset enabled, expandable, intercom, AES encryption, etc.
Svenska kronor1200.00

Level One VOI-7000 IP telephone
Modèle VOI7000
Level One
Key Features   Traditional Phone for digital use Register up to 3 different telephone number User friendly and packed with VoIP and SIP features Enhanced Security features to protect Local Area Networks Cost Effective telephony for long and short distance calls for home, business and enterprise users Plug-and-Play enabled with easy set-up with keypad Hardware Specification

Connection Ports Dual RJ-45 10/100Mbps Ethernet Ports Handset Port RJ-11 port Memory Flash: 2MB SDRAM: 8MB Software Features   Network Protocol: SIP v1/v2, IP/TCP/UDP/RTP/RTCP IP/ICMP/ARP/RARP/SNTP TFTP Client/DHCP Client/PPPeE Client Telnet/HTTP Server DNS Client Codec: G.711: 64kbps (PCM) G.723.1: 6.3kbps/5.3kbps G.726: 16/24/32/40kbps (ADPCM) G.729A: 8kbps (CS-ACELP) G.729B: adds VAD & CNG to G.729A Voice Quality VAD: Voice activity detection CNG: Comfortable noise generator LEC: Line echo canceller Packet Loss Compensation Adaptive Jitter Buffer Security HTTP1.1 authentication for Web setup MD5 for SIP authentication QoS (Quality of Service): ToS Field Configuration Web Browser Keypad Multilingual GUI Register Server (allows for 3 SIP Accounts) Outbound Proxy
Svenska kronor960.00

Linksys SPA-921 IP Telephone
Modèle SPA921
Linksys
Stylish and functional in design, the SPA921 VoIP Phone is ideal for a residence or business using a hosted IP telephony service, an IP PBX, or a large scale IP Centrex deployment. The SPA921 leverages industry leading VoIP technology from Linksys to deliver an upgradeable high quality IP Phone that is unparalleled in features, value, and support. Standard features on the SPA921 include a high resolution graphical display, speakerphone, and a 2.5 mm head-set port. The SPA921 supports one line with two call appearances and provides support for three way conferencing, attended call transfer, and placing a call on hold to answer an incoming call. The line can be configured as a unique phone number (or extension), or can be configured to share a number that is assigned to multiple phones. Comprehensive Interoperability and SIP Based Feature SetBased on the SIP standard, the SPA921 has been tested to ensure comprehensive interoperability with equipment from VoIP infrastructure leaders enabling service providers to quickly roll-out competitive, feature rich services to their customers. With hundreds of features and configurable sevice parameters, the SPA921 addresses the requirements of traditional business users while leveraging the advantages of IP telephony. Features such as easy station moves, presence, and shared line appearances (across local and geographically dispersed locations) are just some of the many advantages of the SPA921. Carrier-Grade Security, Provisioning, and ManagementThe SPA921 uses standard encryption protocols to provide secure remote provisioning and unobtrusive in-service software upgrades. Linksys secure remote provisioning tools include detailed performance measurement and troubleshooting features, enabling network providers to deliver high quality support to their subscribers. Remote provisioning also saves service providers the hassle and expense of managing, pre-loading, and re-configuring customer premise equipment (CPE).   User Guide SPA921 Datasheet SPA921
Svenska kronor1000.00

Linksys SPA-922 IP Telephone
Modèle SPA922
Linksys
Stylish and functional in design, the SPA922 VoIP Phone is ideal for a residence or business using a hosted IP telephony service, an IP PBX, or a large-scale IP Centrex deployment. The SPA922 leverages industry leading VoIP technology from Linksys to deliver an upgradeable high quality IP Phone that is unparalleled in features, value, and support. Standard features on the SPA922 include dual switched Ethernet ports, 802.3af PoE, a high resolution graphical display, speakerphone and a 2.5 mm head-set port. The SPA922 supports one line with two call appearances and provides support for three way conferencing, attended call transfer, and placing a call on hold to answer an incoming call. The line can be configured as a unique phone number (or extension), or can be configured to share a number that is assigned to multiple phones. Comprehensive Interoperability and SIP Based Feature SetBased on the SIP standard, the SPA922 has been tested to ensure comprehensive interoperability with equipment from VoIP infrastructure leaders enabling service providers to quickly roll-out competitive, feature rich services to their customers. With hundreds of features and configurable sevice parameters, the SPA922 addresses the requirements of traditional business users while leveraging the advantages of IP telephony. Features such as easy station moves, presence, and shared line appearances (across local and geographically dispersed locations) are just some of the many advantages of the SPA922. Carrier-Grade Security, Provisioning, and ManagementThe SPA922 uses standard encryption protocols to provide secure remote provisioning and unobtrusive in-service software upgrades. Linksys secure remote provisioning tools include detailed performance measurement and troubleshooting features, enabling network providers to deliver high quality support to their subscribers. Remote provisioning also saves service providers the hassle and expense of managing, pre-loading, and re-configuring customer premise equipment (CPE).   User Guide SPA922 Datasheet SPA922
Svenska kronor1040.00

Linksys SPA-941 IP telephone
Modèle SPA-941
Linksys
This is a SIP based IP phone from Linksys/Sipura. The SPA-941 is suitable for use with SIP based IP PBX systems, or together with a SIP service provider. The SPA-941 phone is the first VoIP Telephone from Linksys since they bought Sipura. The SPA-941 is an affordable, full featured, stylish IP telephone with multiline appearances, graphical display, speakerphone, headset jack and message waiting indicator with a dedicated message retrieval button. Stylish and functional in design, the SPA- 941 can be used in SOHO, enterprise, small to medium businesses together with service offerings including IP PBX, hosted IP telephony and IP Centrex. The SPA-941 is simple to use but as with all Sipura/Linksys products, there are a complete set of configuration options for the advanced user. User Guide SPA941 Datasheet SPA941
Svenska kronor1080.00

Linksys SPA-942 IP telephone
Modèle SPA-942
Linksys
Stylish and functional in design, the SPA942 VoIP telephone is ideal for a residence or business using a hosted IP telephony service, an IP PBX, or a large scale IP Centrex deployment. The SPA942 leverages industry leading VoIP technology from Linksys to deliver an upgradeable high quality IP Phone that is unparalleled in features, value, and support. Standard features on the SPA942 include two active lines, dual switched Ethernet ports, 802.3af PoE support, a high resolution graphical display, speakerphone, and a 2.5 mm head-set port. With a simple software update, the SPA942 is upgradeable to a four line phone. Each line can be independently configured to use a unique phone number (or extension), or can be configured to use a shared number that is assigned to multiple phones. Comprehensive Interoperability and SIP Based Feature SetBased on the SIP standard, the SPA942 has been tested to ensure comprehensive interoperability with equipment from VoIP infrastructure leaders enabling service providers to quickly roll-out competitive, feature rich services to their customers. With hundreds of features and configurable service parameters, the SPA942 addresses the requirements of traditional business users while leveraging the advantages of IP telephony. Features such as easy station moves, presence, and shared line appearances (across local and geographically dispersed locations) are just some of the many advantages of the SPA942. Carrier-Grade Security, Provisioning, and ManagementThe SPA942 uses standard encryption protocols to provide secure remote provisioning and unobtrusive in-service software upgrades. Linksys secure remote provisioning tools include detailed performance measurement and troubleshooting features, enabling network providers to deliver high quality support to their subscribers. Remote provisioning also saves service providers the hassle and expense of managing, pre-loading, and re-configuring customer premise equipment (CPE). User Guide SPA942 Datasheet SPA942
~~Svenska kronor1200.00~~ Svenska kronor1040.00

Linksys SPA-962 IP telephone
Modèle SPA962
Linksys
Stylish and functional in design, the SPA962 VoIP telephone is a must for businesses using a hosted IP telephony service, an IP PBX, or a large scale IP Centrex deployment. The SPA962 leverages industry leading VoIP technology from Linksys to deliver a high quality IP Phone that is unparalleled in features, value, and support. Standard features on the SPA962 include six active lines, dual switched Ethernet ports, 802.3af PoE support, a high resolution color display, speakerphone, and a 2.5 mm head-set port. Each line can be independently configured to use a unique phone number (or extension), or can be configured to use a shared number that is assigned to multiple phones. The power supply for the SPA962 is sold separately and will be required if PoE functionality is not implemented. Comprehensive Interoperability and SIP Based Feature SetBased on the SIP standard, the SPA962 has been tested to ensure comprehensive interoperability with equipment from VoIP infrastructure leaders enabling service providers to quickly roll-out competitive, feature rich services to their customers. With hundreds of features and configurable service parameters, the SPA962 addresses the requirements of traditional business users while leveraging the advantages of IP telephony. Features such as easy station moves, presence, and shared line appearances (across local and geographically dispersed locations) are just some of the many advantages of the SPA962. Carrier-Grade Security, Provisioning, and ManagementThe SPA962 uses standard encryption protocols to provide secure remote provisioning and unobtrusive in-service software upgrades. Linksys secure remote provisioning tools include detailed performance measurement and troubleshooting features, enabling network providers to deliver high quality support to their subscribers. Remote provisioning also saves service providers the hassle and expense of managing, pre-loading, and re-configuring customer premise equipment (CPE).
~~**Svenska kronor2360.00**~~ **Svenska kronor1920.00**

Polycom SoundPoint IP 301
Modèle IP301
Polycom
Built to Polycom’s high standards of quality and usability, the SoundPoint IP 301 offers businesses a cost-effective choice for their IP communication needs.
Svenska kronor1200.00

Polycom SoundPoint IP 320 IP telephone
Modèle IP320
Polycom
Introducing the Polycom SoundPoint 320 two-line SIP phone that delivers superb sound quality as well as a wide range of supported business telephony features. The SoundPoint IP 330 phone, with its dual-port 10/100 Ethernet switch for LAN and PC connection, presents a cost-effective solution for cubicle workers as well as call center operators who use a “hard” phone in conjunction with a “soft” client running on the PC. The SoundPoint IP 320 phone has a single 10/100 Ethernet port, therefore its application targets include common areas, such as lobbies, hallways, and break rooms, as well as various wall-mounted deployments.The Polycom SoundPoint IP 320 phones feature a full-duplex IEEE 1329 Type 1-compliant speakerphone with Polycom's legendary Acoustic Clarity Technology that enables clear, noise-free and echo-free conversations that are as natural as being there.The Polycom SoundPoint IP 320 phones deliver through an intuitive user interface a full feature set encompassing both traditional business telephony features such as call hold, park, pick-up, transfer, and three-way local conferencing, and more advanced capabilities such as shared call/bridged line appearance, support of XHTML applications, and distinctive call treatment.The Polycom SoundPoint IP 320 phones are engineered to make installation, configuration, and upgrades as simple and efficient as possible. The phones’ standard base stand can be reversed to become a wall mount, eliminating the need for a separate accessory. Built-in IEEE 802.3af PoE circuitry and a dual-port Ethernet switch (SoundPoint IP 330 phone only) enable flexible deployment options and savings on cabling expenses.The SoundPoint IP 320 phones have been tested and certified to deliver comprehensive interoperability and extensive feature support with leading SIP-based call control platforms by Broadsoft, Sylantro, Interactive Intelligence/Vonexus, Digium, and other Polycom Technology and Interoperability Partners2. The phones also interoperate with Microsoft ® Live Communications Server (LCS) 2005 for telephony and presence. This broad interoperability offers Polycom customers a choice of supported standards-based call server solutions with which the phones can be deployed.
Svenska kronor1040.00

Polycom SoundPoint IP 430
Modèle IP430
Polycom
The SoundPoint® IP 430 is a two-line desktop IP phone that delivers the outstanding voice quality and smoothness of natural two-way conversations with its full-duplex speakerphone featuring the Polycom Acoustic Clarity Technology. An enterprise-grade IP phone, The SoundPoint® IP 430 is designed to meet the telephony needs of general business users – cubicle workers that conduct a low-to-medium volume of calls – by delivering a robust feature set encompassing traditional telephony features such as call, park, pick-up, hold and transfer, as well as more advanced capabilities such as shared call / bridged line appearance, multiple call appearances, and presence. The SoundPoint® IP 430 offers an easy transition from legacy PSTN systems to the world of IP telephony by providing an intuitive user interface, easy-to-read graphical LCD display, and convenient, one-button access to essential telephony features. With integrated Power over Ethernet circuitry (IEEE 802.3af) and a dual-port 10/100 Mbps Ethernet switch for PC and LAN connection, the SoundPoint® IP 430 offers a choice of powering and cabling options to help reduce cabling expenses and cord clutter. Datasheet IP430
Svenska kronor1440.00

Polycom SoundPoint IP 501
Modèle IP501
Polycom
Enhanced productivity
Svenska kronor1800.00

Polycom SoundPoint IP 601
Modèle IP601
Polycom
The SoundPoint® IP 601 – based on the award-winning SoundPoint® IP 600 – is your best choice for a standards-based Voice over IP telephone. Ideal for users requiring multiple lines and advanced features, the SoundPoint® IP 601 delivers both traditional telephone capabilities and new, converged applications to your business desktop. The SoundPoint® IP 601 provides a powerful, yet flexible IP communications solution, delivering excellent voice quality. Its intuitive user interface offers dedicated, single-button access to common telephony features. The high-resolution display supplies content for call information, multiple languages, directory access, system status, and future applications. The SoundPoint® IP 601 supports advanced functionality, including multiple call- and flexible line appearances, HTTPS secure provisioning, instant messaging, presence, custom ring tones, and three-way local conferencing. When equipped with up to three SoundPoint® IP Expansion Modules, the SoundPoint® IP 601 delivers the advanced call handling capabilities of a high-performance attendant console. Designed to improve productivity of telephone attendants – executive assistants, receptionists, and secretaries – the attendant console allows effective and efficient management and monitoring of a high volume of simultaneous calls. The SoundPoint® IP 601 is easy to install and set up, as it is equipped with a two-port Ethernet switch, built-in auto-sensing Power over Ethernet circuitry, and can be centrally configured from a network-connected server. The SoundPoint® IP 601 offers investment protection because it is software upgradeable in the field. Datasheet IP601
Svenska kronor2400.00

Polycom SoundStation IP Extension Module
Modèle IP-EXT
Polycom
The SoundPoint® IP Expansion Module for the SoundPoint® IP 601 is an optimal solution for telephone attendants – receptionists, administrative assistants, secretaries, and other “power users” who manage and monitor multiple simultaneous telephone calls on a daily basis. The SoundPoint IP Expansion Module augments the user interface of the SoundPoint® IP 601 with a high-resolution graphical LCD and 14 multifunctional line keys that can be set up as a line registration, call appearance, or a speed-dial. Equipped with a high-performance attendant console based on the SoundPoint® IP 601 and up to three SoundPoint® IP Expansion Modules, the telephone attendant can reduce the number of lost customer calls, shorten transaction times, and increase the accuracy of call routing by promptly accepting, accurately screening, efficiently dispatching, and effortlessly monitoring calls. The SoundPoint® IP Expansion Module provides an easy transition from traditional PBX features and functionality into the world of Voice over IP with its large, high-resolution LCD, 14 illuminated multifunctional line keys, and user-friendly call visualization similar to that on the SoundPoint® IP 601. The SoundPoint® IP Expansion Module is a true “plug-and-play” device that requires no set-up as power and signaling are provided by the host SoundPoint® IP 601 phone. No extra cables or power supplies are required to operate the Expansion Module. Just snap it on – and it works. Like all Polycom VoIP products, the SoundPoint IP Expansion Module is standards-based and fully interoperable with Polycom Partners' IP telephony server solutions. Datasheet Expansion Module
Svenska kronor2000.00

Polycom SoundStation IP4000
Modèle IP4000
Polycom
With improved audio quality over the original Polycom SoundStation IP 3000 conference phone, the SoundStation IP 4000 SIP voice conferencing unit is the answer for organizations that are ready for the benefits and versatility of a SIP enabled business.Designed for offices or small to medium sized conference rooms, the SoundStation IP 4000 SIP provides remarkable room coverage. You can speak naturally from up to 10-feet away from a microphone and still be heard clearly on the far end of the call. Need coverage for a larger room? The optional extension microphones offer an increased pickup for larger rooms. Plus, with gated microphone technology, echo and background noise is almost entirely eliminated. Datasheet IP4000
~~Svenska kronor6960.00~~ **Svenska kronor5840.00**

Siemens OpenStage 20 IP telephone
Modèle SOS20
Siemens
Product description will be available soon!
Svenska kronor1680.00

Siemens OpenStage 40 IP telephone
Modèle SOS40
Siemens
Product information will be available soon!
Svenska kronor2320.00

Siemens OpenStage 60 IP telephone
Modèle SOS60
Siemens
Product information will be available soon!
Svenska kronor3120.00

Siemens OpenStage 80 IP telephone
Modèle SOS80
Siemens
Product description will be available soon!
Svenska kronor5360.00

Snom 300 IP Telephone
Modèle 300IP
Snom
For effective everyday work, the snom 300 provides all relevant office functions such as choice of trunk line, status display, group lines, the engaged option or picking up calls.
Svenska kronor1040.00

Snom 320 IP telephone & Headset HS-MM2
Modèle 320SIPH
Snom
A 100-number call memory, 100-number onboard address book (to which data may easily be uploaded), custom call blocking, configurable/downloadable ring-tones, auto-answer mode, DND and other sophisticated features insure convenience and productivity. And the 320's built-in web server supports even simpler end-user configuration, screen dialing, and access to call history.The snom 320 is remote-manageable and firmware-upgradeable, uniquely easy to install, and largely self-configuring. Broad codec support and full compatibility with current SIP recommendations insures interoperability; support for STUN (for NAT traversal), ENUM (for dialed-number resolution) and other state-of-the-art features enables flexible deployment, behind local proxies, IP PBXs or hosted VoIP services.The snom 320 supports the security standard SRTP - a current specification from the Internet Engineering Task Force (IETF) for protection against eavesdropping - and TLS for protection against sniffing of signaling and authentication data.By limiting need for external conference bridges/media server capacity or use of conference services for routine multiparty calls, the snom 320's built-in three-party conference bridge helps limit total cost of ownership, while also insuring high audio quality and low latency.Features:two-line display with graphical field47 keys, 13 LEDs12 programmable function keysSpeakerphoneDual Ethernet connectionPower over EthernetHeadset connectionSIP RFC3261Security: SIPS/SRTPSTUN, ENUM, NAT, UPnP, ICECompression: G.723.1 and othersNational Language Support Manual Snom 320 Quick Installation Snom 320
Svenska kronor2160.00

Snom 320 SIP Telephone
Modèle 320 SIP
Snom
A 100-number call memory, 100-number onboard address book (to which data may easily be uploaded), custom call blocking, configurable/downloadable ring-tones, auto-answer mode, DND and other sophisticated features insure convenience and productivity. And the 320's built-in web server supports even simpler end-user configuration, screen dialing, and access to call history.The snom 320 is remote-manageable and firmware-upgradeable, uniquely easy to install, and largely self-configuring. Broad codec support and full compatibility with current SIP recommendations insures interoperability; support for STUN (for NAT traversal), ENUM (for dialed-number resolution) and other state-of-the-art features enables flexible deployment, behind local proxies, IP PBXs or hosted VoIP services.The snom 320 supports the security standard SRTP - a current specification from the Internet Engineering Task Force (IETF) for protection against eavesdropping - and TLS for protection against sniffing of signaling and authentication data.By limiting need for external conference bridges/media server capacity or use of conference services for routine multiparty calls, the snom 320's built-in three-party conference bridge helps limit total cost of ownership, while also insuring high audio quality and low latency.Features:two-line display with graphical field47 keys, 13 LEDs12 programmable function keysSpeakerphoneDual Ethernet connectionPower over EthernetHeadset connectionSIP RFC3261Security: SIPS/SRTPSTUN, ENUM, NAT, UPnP, ICECompression: G.723.1 and othersNational Language Support Manual Snom 320 Quick Installation Snom 320
Svenska kronor1600.00

Snom 360 IP telephone
Modèle 360 IP
Snom
The snom 360 was designed for maximum productivity and efficiency in the everyday business environment. Dedicated keys provide you with direct access to the functions for audio and call control, and context-sensitive menus offer you the additional functionality that you may need at any given moment. The graphical display can be tilted for optimum reading angle.Additional sophisticated call control features, full call detail, configuration options and online help can be accessed via browser from the attached PC.Customized ring tones can be downloaded without any hassle from the web server - including, of course, that catchy ring tone that you like so much. Incoming calls can be marked with special ring tones to indicate the destination of the call.12 programmable keys can be used to customize the functionality according to your own specific needs. The LED behind the corresponding function keys can show you whether or not your colleagues are currently on a call. And of course your colleagues can see whether your line is free or not.To spare you the annoyance of unwanted invasions on your speech data, the snom 360 supports the security standard SRTP - a current specification from the Internet Engineering Task Force (IETF) for protection against eavesdropping and the stealing of data.With SIP (Session Initiation Protocol) you are ensuring your own personal independence. Most vendors are tipping SIP to be the communication protocol of the future. SIP components can be combined into a complete system without the need to be tied to a single provider. Features:Tiltable graphical display (128x64 pixels)47 keys, 13 LEDs12 programmable function keysSpeakerphoneDual Ethernet connectionPower over EthernetHeadset connectionUp to two additional keypads (each with 42 programmable function keys)SIP RFC3261Security: SIPS/SRTPSTUN, ENUM, NAT, UPnP, ICECompression: G.723.1 and othersNational Language SupportAsian Language Support Manual Snom 360 Quick Installation Snom 360
Svenska kronor1920.00

Snom 360 IP Telephone & Expansion Module
Modèle 360IPEX
Snom
The snom 360 was designed for maximum productivity and efficiency in the everyday business environment. Dedicated keys provide you with direct access to the functions for audio and call control, and context-sensitive menus offer you the additional functionality that you may need at any given moment. The graphical display can be tilted for optimum reading angle. Additional sophisticated call control features, full call detail, configuration options and online help can be accessed via browser from the attached PC.Customized ring tones can be downloaded without any hassle from the web server - including, of course, that catchy ring tone that you like so much. Incoming calls can be marked with special ring tones to indicate the destination of the call.12 programmable keys can be used to customize the functionality according to your own specific needs. The LED behind the corresponding function keys can show you whether or not your colleagues are currently on a call. And of course your colleagues can see whether your line is free or not. To spare you the annoyance of unwanted invasions on your speech data, the snom 360 supports the security standard SRTP - a current specification from the Internet Engineering Task Force (IETF) for protection against eavesdropping and the stealing of data. With SIP (Session Initiation Protocol) you are ensuring your own personal independence. Most vendors are tipping SIP to be the communication protocol of the future. SIP components can be combined into a complete system without the need to be tied to a single provider. Manual Snom 360 Expansion Module
~~Svenska kronor2920.00~~ **Svenska kronor2720.00**

Snom 370 IP telephone
Modèle S370
Snom
With the large graphical, high-definition display, the snom 370 offers an improved and extended presentation of call lists, address books and caller information. Caller information can be customized easily through XML to depict the information the user wants displayed. "This means for users and administrators a decisive ease of information and configuration management" Through the mini browser, users can have direct access to their own applications via the display screen. Users can not only customize the design of the display but also view news tickers and other information as well as access central or public phone directories. The snom 370 supports several audio devices simultaneously, for example, it is possible to use the handset, headset, and loudspeaker concurrently. The SIP-based VoIP telephone offers with its expanded memory capacity in addition to all necessary office functionality such as choice of trunk line, status indicator, group lines, transfer, call-pickup or conferencing (3-way conference bridge) more scope for individual functions and applications. The expanded memory capacity also enables the depiction of graphics and high resolution pictures to show the status of contacts (e.g., busy, on-line, off-line) similar to Instant Messenger. It is also possible to play music and media files. snom continues to focus on the highest security features necessary for businesses, especially sensitive business environments, such as hospitals, finance, insurance or military. To avoid problems with unwanted violations of audio data, the snom 370 supports the current security standards such as TLS (Transport Layer Security), SRTP and SIPS. Additionally, the 370 supports VPN for the maximum protection against electronic eavesdropping and data theft. High audio quality, ease of use, security and interoperability make the snom 370 highly suitable for SMEs, home offices, private users or ISP applications.

Most important features
Tiltable, high-definition graphical display (240 x 128 Pixel)
Large LED (red) for incoming calls
47 keys, 13 LEDs
12 programmable function keys
Speakerphone
Dual Ethernet connection
Power over Ethernet
Support for several audio devices
Additional keypads with 42 programmable function keys
SIP RFC3261
Security: SIPS/SRTP, TLS
STUN, ENUM, NAT, ICE
Codecs: G.711, G.729A, G.723.1, G.722, G.726, GSM 6.10 (Full rate)
National Language Support
Svenska kronor2440.00

Thomson Speed Touch 2020 IP Telephone
Modèle T2020
Thomson
The SpeedTouch 2020 offers a high-quality IP solution at a cost effective price making Voice over IP (VoIP) more accessible for businesses.The SpeedTouch 2020 is a VoIP phone that is SIP compliant and can be used with any SIP compliant PBX, Softswitch or IP Centrex solutions of the market. It addresses the need for an cost effective IP solution, helping to make VoIP a viable, attractive option for small or large offices environment. It can be powered over the Ethernet cable and features a second Ethernet port, eliminating the need for a separate Ethernet link and dedicated AC power outlet.
Svenska kronor720.00

Thomson Speed Touch 2030 IP Telephone
Modèle T2030
Thomson
The SpeedTouch 2030 offers a high quality IP solution at a cost effective price making Voice over IP (VoIP) more accessible for businesses. The SpeedTouch 2030 is a VoIP phone that is compliant with SIP/MGCP and can be used with any SIP compliant PBX, Softswitch or IP Centrex solutions of the market. It addresses the need for a cost effective IP solution, helping to make VoIP a viable, attractive option for small or large offices environment.
Svenska kronor1040.00

Thomson SpeedTouch 2022 IP telephone
Modèle T2022
Thomson
The Thomson ST2022 offers a high-quality IP solution at a cost effective price making Voice over IP (VoIP) more accessible for businesses. The Thomson ST2022 is a VoIP phone that is compliant with SIP/MGCP and can be used with any SIP compliant PBX, Softswitch or or IP Centrex solutions of the market. It addresses the need for a cost effective IP solution, helping to make VoIP a viable, attractive solution for small and large ofices environment.   At a Glance: 144x32 full graphic display for 3 lines + icons Connectivity: Integrated 2 port 10/100 Ethernet switch with PoE and VLAN tagging Phone services: 2 lines, Call Forward, Call transfer, Call Hold, Call Waiting Handsfree Alert (missed calls, Message Waiting Indicator) Phone book: Local (100 entries), Remote (HTTP XML) Caller ID Display Automatic Callback 3-ways Conferencing Downloadable rings Multiple power options: power over Ethernet 802.3af and external power supply adapted (included) VoIP Standard: SIP or MGCP Voice compression standards: G.711, G.723, G.729ab IP addressing: Static or dynamic IP coniguration (integrated (DHCP client) Quality of Service: ToS DiffServ, 802.1p/Q, VAD, CNG, Packet Loss Compensation, Adaptative Jitter Buffer Web browser interface for coniguration and Firmware (admin & user mode) Automatic provisioning system (DHCP, TFTP/HTTP) 7 languages
Svenska kronor1160.00

Wi-Fi Telephones

Linksys WIP330 Wi-Fi telephone
Modèle WIP330
Linksys
The Linksys WIP330 Wireless-G IP Phone enables high-quality voice over IP (VoIP) service through a Wireless-G network and high-speed Internet connection. Connect at home, your office, or at a public hotspot, and make low-cost phone calls through your Internet Telephony Service Provider. The handset features peer-to-peer dialing, speed dial, 3-way conferencing, call waiting, call transfer, and call forward, mute, hold and selectable ringtones. The large, full-color high resolution LCD display features an intuitive user interface enabling users to easily and quickly configure the handset when traveling within range of other Wireless-G networks so you can make VoIP calls wherever you go. You can also surf the Internet with the built-in web browser and can even receive live video from anywhere in the world and view it right on your WIP330 Wireless-G IP Phone when you access any web camera, like the WVC54GC Wireless-G Internet Video Camera, also from Linksys! Handset stores the last 20 call history records and can save 250 phone book entries. Personalize your phone with a selection of ringtones that reflects your style. Get the value of low-cost VoIP service while unleashing the full potential of Wireless-G connectivity with the Wireless-G IP Phone from Linksys. User Guide Linksys WIP330
Svenska kronor2400.00

LIP TA100 wi-fi telephone
Modèle LIPTA100
Lip Wi-Fi
LIP’s model TA100 allows you to chat and make free Internet calls or low cost regular calls. It works wherever there is an open wireless broadband connection; offices, residential areas, airports, cafés, hotels and public hot spots. No laptop, headset or cables needed.   TA 100 is a small and light weight phone with high audio quality and an extraordinary talk time of ten hours and up to 250 hours stand-by time. Smart navigation and a simple set-up make it easy to use and easy to set up. All functions can be configured on your phone by the operator before delivery. TA100 enables you to make encrypted calls over the radio link, handle your contacts and send instant messages, beside the traditional functions of mobile phones.

The built-in WiFi includes all common features such
as DHCP, authentication and encryption.
With its flexible open source Linux platform LIP’s WiFi
phones can be designed to fit most VoIP software
and technical standards, as well as specific industrial
design requirements.
LIP’s TA100 is the perfect product for the mobile
generation with high demands on comfort, flexibility
and low cost communication.
Product Model: TA100 – Carbon/SimplyWiFi

Description
Current version
Operating system
Requirements
File size

Upgrade Tool
1.0.7
Microsoft Windows XP
.Net 2.0 installed
266KB
[Download]

Bootloader Tool
1.0.3
Microsoft Windows XP
The phone has to have 0.11.8 software version installed
219KB
[Download]

Full installer (Upgrade tool + Phone software)
Upgrade 1.0.7 Software 1.0.0
Microsoft Windows XP
.Net 2.0 installed
5.74MB
[Download]

Svenska kronor1520.00

Siemens Enterprise optiPoint WL2 professional S WLAN-Telephone
Modèle OPWL2
Siemens
The optiPoint WL2 professional has a backlit color display and offers easy access to a wide range of features. Additional benefits such as an internal phonebook, integration of polyphonic ringtones, voice dialing and access to your company phone directory (LDAP) make the optiPoint WL2 professional the ideal wireless solution.As one of the best VoWLAN company solutions in its class, the optiPoint WL2 professional offers an unrivaled level of voice quality, a range of security features that surpasses the industry standard, and an easy-to-operate user interface. It also supports the IEEE standards for wireless networking (802.11b) and 54 Mbit/s data transmission (802.11g).
Svenska kronor2320.00

Siemens Gigaset SL75 WLAN
Modèle SL75WLAN
Siemens
Gigaset VoIP combines the usual comfort of a cordless phone with the advantages of internet technology, such as accessing messenger services.Gigaset SL75 WLAN let customers enjoy internet phoning from home and private WLAN or open hotspots. Highlights:- VoIP handset for internet calls at home and in private WLANs or open hotspots - Send and receive e-mails without a running computer - Integrated digital camera - Instant messaging (buddy list, chat) and presence services Features- Handset with colored graphical display (6 lines, 128x128 pixel resolution, 4096 colors (65k in future), 3,1cm x 3,1cm)- Display with calling animation signalizes incoming call and CLIP in Display- Intuitive user prompting with optimized MMI- Illuminated MWI key- Handsfree talking key- Status display stand-by-mode: date, time, battery status, RF-signal-strength, connected access point- Status display during talk: battery status, RFsignal-strength, talk-time, off-hook sign, CLIP- Key-Pad with hard-keys (4-way button, 3x4 (0,1..9, #, *), 4 additional keys for MWI, on-hook, off-hook, hands-free, 2 additional programmable keys for applications and 2 separated soft-keys Telephone functionality- Missed Call List- Voice dialing- Hands free talking- List of all received calls- Dialing via numbers, SIP-URI and IP-address (direct IP-call)- Internal Call- Dialing preparation (input of phone numbers before seizure) with correction possibility- Redialing of 10 last different dial numbers- Automatic number redialing- Speed dialing via keys (2 - 9)- Support of Provider Services (e. g. net-answering machine) - dependent on provider requirements (tbd.)- Several talks in parallel (max. 2 per handset); exact number depending on broadband-connection; e. g. 3 talks with G.729ab/128 kbit/s upstream or 6 talks with G.729ab/256 kbit/s Address book- Rich address book for 200 contacts (each contact may contain the following entries: name, first name, 3 numbers (home, office, mobil: each number or SIP-URI or IP address), e-mail, department, street, PLZ, city, department)- CLIP is replaced by Telephone book entry Audio- Up to 16 standard polyphonic ringer tones (volume adjustable)- Loadable ringer-tones- Volume regulation (8 steps)- CLIP dependent ringer tones- Size: 132x52x22 mm- Range indoors up to 30 m- Range outdoors up to 100 m
Svenska kronor1640.00

UTStarcom F3000 Wi-Fi telephone
Modèle F3000
UTStarcom
The UTstarcom F1000 wireless IP phone combines the power of Voice Over IP technology with the mobility of wireless networking. Wireless Voice Over IP is a revolutionary step for VoIP technology because it extends the reach and capabilities of your VoIP network. UTstarcom has become a respected leader the mobile VoIP industry by releasing quality handsets that use the flexible SIP protocol at an affordable cost. UTstarcom gained an innovative reputation in the marketplace by being one of the first manufacturers to release a 802.11g wireless IP phone which drastically increased sound quality and compatibility with modern wireless (wifi) networks.The UTstarcom F1000 provides consumers a new cost effective way to communicate, with great features such as 3-way Calling, Call Waiting, Call Transfer and many popular features.   Datasheet F3000 Manual F3000
~~Svenska kronor2160.00~~ **Svenska kronor2040.00**

Voip Fax

Sagem PhoneFax 43A
Modèle S43A
Sagem
INTERNET PHONE & FAX
Dual line: Internet (VoIP) and PSTN telephone & fax
Simple switching between VoIP and PSTN
Voice Codes: G.711, G.729
Protocols: SIP, RTP, DHCP, STUN, FTP
No computer (PC) needed
Reliable internet fax with T.38 standard
WLAN optional
Quality of service (QoS)*: ToS, DiffServ
Socket for connection of analogue telephone
INK-FILM FAX
9.600 bps modem speed, auto fall back
Horizontal resolution: 8 pixels/mm
vertical resolution standard: 3.85 lines/mm
vertical resolution fine: 7.7 lines/mm
Page header, automatic redial
Sending report with reduced copy of the first page
Backlit display
Printing method: thermal transfer
10 page automatic document feeder (ADF)
Predefined fax forms
Sudoku printing with solutions (easy – medium – tough – devil)
Timer function (send later)
Up to 50 page memory
Group broadcasting
Paper tray up to 50 sheets
Cordless phone management
TELEPHONE
50 name dial entries
10 ringer melodies
Caller identification (CLIP)*
Distinctive ring tones**
6 speed dial keys
Alphabetical keyboard

On hook dialing
Handsfree speaking
Chain dialling
Redial of the last 10 numbers
Outgoing call log print
COPIER
Multicopy up to 15 pages
Fine-/ photo-resolution
Zoom-function (enlarge/reduce)
16 grey scales
* Depends on the availability of the service   on the local operators network ** Depends on availability of caller identification
Svenska kronor1600.00

Sagem PhoneFax 49A
Modèle S49A
Sagem
      Dual Line: Internet + Fixed Line Optional WLAN Connectivity SIP Compatible No PC Required Internet Phone & Fax - Dual line: Internet (VoIP) and PTSN telephone & fax - Simple switching between VoIP and PSTN - SIP standard supported - No computer (PC) needed - Reliable internet fax with T.38 standard - WLAN ready Ink-film plain paper fax- 9.600 bps modem speed (auto fall back) - Sending report with reduced copy of the first page - 10 page automatic document feeder (ADF) - Paper tray up to 50 sheets - Backlit display - Timer function (send later) - Predefined fax forms - Group broadcasting - Cordless phone management Telephone- 200 name dial entries - 10 ringer melodies - Caller identification (CLIP)* - Distinctive ringtones** - 10 short dial keys - On hook dialing - Handsfree speaking - Redial of the last 10 numbers -Outgoing call log printDECT cordless handset- Colour graphic display (256 colours) - GAP compatible (up to 6 handsets) - Up to 300m range - 120 hours stand by time, 10 hours talk time - 12 HIFI ringtones/ 11 standard ringtones - Access to fax phonebook - 3 x AAA NiMH - Backlit display - INTERCOM & call transferAnswering machine- Recording time up to 30 minutes - Current call recording Copier- Multicopy up to 15 pages - Zoom-function (enlarge/reduce) - 16 grey scales * Depends on the availability of the service   on the local operators network ** Depends on availability of caller identification
Svenska kronor2360.00

Skype Telephones

Linksys CIT200 Skype telephone
Modèle CIT200
Linksys
You don't have to sit in front of your computer to talk to your friends with Skype™ USB Base Station connects to your PC and puts your Skype contact list on the handset's screen -- just choose and talk Handset features illuminated Caller ID screen, speakerphone, call waiting, and intercom to other (optional) handsets Also supports SkypeOut dialing™ incoming SkypeIn™ calls and Skype VoiceMail Get cordless handset convenience when you use Skype! Now you don't have to sit in front of your computer when you want to talk to your friends. The Linksys Cordless Internet Telephony Kit comes with the Handset and its charger, and a USB Base Station that plugs into your computer. With the included Skype program running on your PC, the Handset can read and display your contact list on its built-in illuminated display. Just choose who you want to talk to from the menu, press a button, and get ready to talk. Of course, the Handset supports SkypeOut dialing as well as your Skype contact list. It rings when you have a Skype or SkypeIn call coming in, and shows the caller's ID on the screen. The Handset also supports call waiting and Skype's optional Voicemail service. When you're on a call, you can hit the Mute button for privacy, or use the speakerphone function to let everyone join in. You can even use the phone as an intercom -- each Base Station supports up to four Handsets. In short, the Linksys Cordless Internet Telephony Kit gives you all the functionality of a regular phone, through Skype. You'll get the convenience of a cordless phone and all its features, plus the high-quality voice reception that you get from Skype on your PC, without being tied to your computer. Datasheet CIT200 Manual CIT200 Quick Installation CIT200
Svenska kronor792.00

Linksys CIT300 Skype teöephone
Modèle CIT300
Linksys
Get cordless handset convenience when you use Skype! Now you don't have to sit in front of your computer when you want to talk to your friends. The Linksys cordless Dual-Mode Internet Telephony Kit comes with the handset, charger, and a USB Base Station that plugs into your computer. Dual-Mode capability lets you choose whether to make calls over a standard telephone line or with the Skype application running on your PC. Just select who you want to talk to from your contact list on your Handset, press a button, and get ready to talk. Of course, the Handset supports SkypeOut dialing as well as your Skype contact list. It rings when you have a Skype or SkypeIn call coming in, and shows the caller's ID on the screen. The Handset also supports call waiting and Skype's optional Voicemail service. When you're on a call, you can hit the Mute button for privacy, or use the speakerphone function to let everyone join in. In short, the Linksys Dual-Mode Cordless Internet Telephony Kit allows you to choose to make calls to other Skype users or over your existing standard telephone line. You'll get the convenience of a cordless phone and all its features, plus the high-quality voice reception that you get from Skype on your PC, without being tied to your computer. The Linksys CIT300 corresponds basically to the CIT200. Additionally it has a build- in PSTN port. This offers to do calls weather through Skype or the conventional telephone network. Furthermore the CIT300 supports “Skype voicemail”. The standby time is up to 120 hours, the talk time up to 10 hours. Main features - Skype through a common DECT phone - PSTN port - “Skype voicemail” supportTechnical specifications - Standards: -- Connection Port: - - Protocol: DECT - Audio CODEC: - - Management: via keypad - Fax support: no - Dimensions: 47,37 mm x 34,60 mm x 132 mm (WxDxH),   0,13 kg (phone)
Svenska kronor920.00

Polycom Communicator for Skype
Modèle PCS
Polycom
Polycom Communicator gives you the ultimate hands-free Skype experience. Based on the same technology used in Polycom’s legendary line of triangular SoundStation conference phones, the Skype-certified Polycom Communicator enables crystal-clear, natural conversations when using Skype. Enjoy the freedom of not wearing your headset for hands-free Skype calls, or plug into the built-in stereo headphone port for private conversations. The Polycom Communicator delivers high-fidelity wideband voice quality that sounds like you are in the same room with the people you’re calling. Two high-quality microphones provide excellent range for group conversations with up to four participants. Polycom’s Acoustic Clarity Technology eliminates echoes and feedback, maximizing your Skype experience. Features: High Fidelity Wideband Voice Quality Two High Quality microphones Skype Certified Built-In stereo headphone port Connects and Powers through USB
Svenska kronor960.00

Smartphones with Voip

Pirelli Discus DP-L10, Dual-mode Wi-Fi / GSM telephone ** Unloc
Modèle DP-L10
Pirelli Broadband
Tri-band GSM (900/1800/1900 MHz)
your customers can use the phone both under domestic and foreign mobile networks GPRS
Your customers can use the standard mobile services for SMS and MMS they are used to when in mobility GCF (Global Certification Forum) certified
Your can offer your customers freedom to choose subscription to any licensed GSM mobile operator, with no issue of interoperability SMS and MMS
Maintains the user experience for established data services the mobile users are accustomed to. WiFi802.11b/g
The phone works with any standard WiFiAccess Point
The phone does not cause reduction of throughput for data clients (PCs), as happens with 11b only phones
The phone can be used within outdoor WiFiHotspot networks Up to 5 WLAN use profiles:
Up to 5 user defined WLAN profiles can be configured (Home, Office, Hotspot, User1, …), each with pre-defined WLAN settings: the phone automatically searches for available VoIPWLAN service profile -> nomadicity
Intelligent call routing:
Emergency call numbers are automatically sent over GSM, adding a key functionality missing in standard broadband VoIPservices
User can manually select for each number whether to send it via GSM or WLAN network SIP 2.0 VoIPprotocol
The phone is interoperable with leading softswitchplatforms
Different than UMA phones, the PBS DP is suitable for utilization in business environments requiring handover between APs Remote provisioning (OMA OTAP protocol)
Zero touch configuration
No user manual configuration is needed for service provisioning and activation
Software upgrade can be managed by the network WAP 2.0 and E-mail (both via GPRS and WiFi)
When under the coverage of WiFiaccess points, your customers can, (which is not possible with mobile phones):
Experience fast connectivity in browsing via WAP and reading e-mail attachments
save on the cost of expensive long GPRS connection sessions Personal PC utility:
Your customers can download /upload via USB their personal files (photos, wallpapers, ringtones) Your customers can synchronize e-mail and contacts between PC and phone Extremely low SAR:
The Pirelli DP in WiFimode achieves SAR values up to ten times lower than those produced by GSM phones -> your service can create both perceived and actual higher safety for the user in terms of electromagnetic energy radiation towards the body Main features
Protocol: GSM 900/1800/1900 MHz + WiFi(802.11b/g)
Size: 106 x 46 x18 mm
Weight: 95 g
Form Factor: Bar type
Display: 65K CSTN 1.5", 128x128 pixel
Camera: 300K pixelVGA
OMA OTAP for SIP and Wi-Fisettings
Talk time: 240min under GSMmode
Standby time: 180hr, GSM mode150hr, WiFimode w/o periodic scan for roaming
Battery: 1050mAh

Antenna: Internal
I/O Interface: mini-USB
Accessories:
standard: travel charger, battery, manual, mini-USB cable
options: cradle charger, earphone, PC utility
2Data services: • WAP 1.x/2.0• SMS• MMS Client1.0• Email Client SMTP and POP3
WiFiVoIP•Security: WEP64, WEP128, WPA-PSK-TKIP, WPA2-PSK-AES•Vocoder: G.711, G. 726, G.729
Seamless roaming within WiFiwhen call active.
500 entries phonebook
60 entries for dialed, missed and received calls (20 for each)
10 pre-installed wall paper; 40 pre-installed polyphonic ring tones
5 pre-installed power on/off animation; 2 pre-installed Games
Screen protection
Utilities-Calculator -Currency -Stop Watch -Count Down-Scheduler -Calendar-World time -Alarm
http://www.youtube.com/watch?v=GPtYZWNYjm0
~~**Svenska kronor2200.00**~~ **Svenska kronor2080.00**

Analogue Telephone Adapter

Linksys PAP2T analogue adapter
Modèle PAP2T
Linksys
 The Linksys Phone Adapter enables high-quality feature-rich telephone service through your cable or DSL Internet connection. Just plug it into your home router or gateway and use the two standard telephone jacks to connect your existing phones or fax machines. Each phone jack operates independently, with separate phone service and phone numbers - like having two phone lines. With an appropriate Internet telephone service provider, you'll get clear telephone reception and reliable fax connections, even while using the Internet at the same time for normal data operations.With Internet telephony, along with low domestic and international phone rates, an impressive array of special phone features are available. Choose your preferred free local dialing US area code, regardless of where you live. Or add a virtual phone number in any area code, forwarded to your Internet phone. You can even add a toll-free number. The Linksys Phone Adapter is compatible with these and all of the other special telephone features that are available from your telephone service provider, such as Caller ID, Call Waiting, Voicemail, Call Forwarding, Distinctive Ring, etc.Let the Linksys Phone Adapter turn your existing Internet connection into a high-quality high-value telephone service. Datasheet PAP2T Manual PAP2T Quick Installation PAP2T
~~Svenska kronor480.00~~ **Svenska kronor376.00**

Linksys SPA-2102 analog
Modèle SPA-2102
Linksys
Inexpensive, easy-to-install and simple-to-use, the SPA-2102 connects standard telephones and fax machines to IP-based data networks. IP telephony service providers and enterprise users can offer traditional and enhanced communication services via the customers' broadband connection to the Internet or Local Area Network (LAN). The SPA-2102 features two POTS (Plain Old Telephone Service) ports for connection to existing analog telephones, fax machines, PBX and key system communication platforms. The SPA-2102 includes an Ethernet interface for connecting to a home or office PC (LAN) as well as an Ethernet connection to the broadband modem or router (WAN).Each SPA-2102 service line can be independently configured via software controlled by the service provider and/or the end user. Installed by the end user and remotely provisioned, configured and maintained by the service provider, each SPA-2102 converts voice traffic into data packets for transmission over an IP network. Upstream voice traffic can take priority over less time sensitive data traffic to provide quality of service (QoS) treatment to VoIP phone calls.Stylish and compact in design, the SPA-2102 can be used in consumer and business IP telephony service offerings including full-featured IP Centrex. The SPA-2102 delivers clear sounding voice and reliable fax calling through its implementation of internationally recognized standards for voice and data networking.
~~**Svenska kronor680.00**~~ **Svenska kronor560.00**

Linksys SPA3102 analogue adapter
Modèle SPA3102
Linksys
The SPA3102 features the ability to connect standard telephones and fax machines to IP-based data networks with the additional benefit of an integrated connection for legacy telephone network "hop-on, hop-off" applications. SPA3102 users will be able to leverage their broadband phone service more than ever by automatically routing local calls from mobile phones and land lines over to VoIP service providers and vice versa. If power is lost to the unit or Internet service is down, calls can be redirected to a traditional carrier via the FXO interface. A user calling from a mobile phone or land line will be able to reduce and even eliminate international and long distance telephone charges by first calling their SPA3102 via a local telephone number. The advanced authentication and call routing intelligence programmed into the SPA3102 will route the call via the Internet to the far end destination. In addition, when using the SPA3102 at the far end, VoIP calls placed to that location can be either answered or further processed and routed on as a local call to any legacy land line or mobile phone. The SPA3102 supports one RJ-11 POTS (Plain Old Telephone Service) FXS port to connect an existing analog phone or fax machine. The SPA3102 also supports one PSTN FXO port to connect to a Telco or PBX circuit. The SPA3102 includes 2 100BaseT RJ-45 Ethernet interfaces to connect to a home or office LAN, as well as an Ethernet connection to a broadband modem or router. The SPA3102 FXS and FXO lines can be independently configured via software controlled by the service provider or the end user. Installed by the end user and remotely provisioned, configured and maintained by the service provider, each SPA3102 converts voice traffic into data packets for transmission over an IP network. Compact in design, the SPA3102 can be used in consumer and business VoIP service offerings including a full-featured IP Centrex environment. The SPA3102 uses international standards for voice and data networking for reliable voice and fax operation. HighlightsToll Quality Voice and Carrier-Grade Feature SupportThe SPA3102 delivers clear, high-quality voice communication in diverse network conditions. Excellent voice quality in a demanding IP network is consistently achieved via our advanced implementation of standard voice coding algorithms. The SPA3102 is interoperable with common telephony equipment like voicemail, Fax, PBX, and interactive voice response systems.Large-Scale Deployment and ManagementThe SPA3102 offers all the key features and capabilities which service providers can provide customized VoIP services to their subscribers. The SPA3102 can be remotely provisioned and supports dynamic, in-service software upgrades. A secure profile upload saves providers the time, expense, and hassle of managing and pre-configuring or re-configuring customer premise equipment (CPE) for deployment.Ironclad SecurityLinksys understands that security for end users and service providers is a fundamental requirement for a solid, carrier-grade telephony service. The SPA3102 supports secure, standard encryption-based methods for communication, provisioning and servicing.
~~Svenska kronor720.00~~ **Svenska kronor680.00**

Linksys SPA400 analogue adapter
Modèle SPA400
Linksys
The SPA400 features the ability to connect up to four (4) standard analog telephones lines to a Linksys Voice System (LVS) VoIP network and includes the additional benefit of an integrated voicemail application. The SPA400 utilizes multiple analog phone lines and automatically routes calls to and from your existing PSTN telephone service. Designed to be implemented with the LVS IP Telephony System, the SPA400 enables cost-conscience business users to leverage the high value features generally found only on more expensive PBX equipment. The SPA400 also includes an integrated voicemail application supporting up to 32 voicemail accounts or boxes with customized greetings, providing LVS users the ability to receive and playback voicemail messages from their Linksys IP Phones. The SPA400 supports four (4) RJ-11 FXO ports to connect to the PSTN and also includes one 10/100 BaseT RJ-45 Ethernet interface that is used to connect to the local IP network. Compact in design, the SPA400 is perfectly suited for a small business in providing reliable operation for both VoIP and PSTN voice communication service. Installed by the service provider, reseller, or end user, the SPA400 enables transparent PSTN gateway functionality for a VoIP telephony network and is a key element in building a solid business communications system.
Svenska kronor1680.00

Voip Router

Draytek 3300V router
Modèle 3300V
Draytek
 The Vigor3300V provides optional 4/8 port integrated FXO or FXS VoIP module to connect to PBX or phone to minimize voice operation cost. It provides similar voice quality, reliability as the traditional telephones and security conversion to business users. With T.38 fax over IP support, customers can provide real-time, secured, reliable and free fax service. The Vigor3300 series delivers enterprise-level security router services for small and medium sized companies. These products combine Virtual Private Network (VPN) 200 tunnels with robust firewall, load balance, web content filtering, QoS, bandwidth guarantee, management, secure administration, and high availability capabilities to meet the increasing demand for reliable, security broadband business services. The enterprise-level security router provides further IPSec VPN to achieve the objectives of high scalable, confidential e-business, branch office interconnection and remote access applications. By using a hardware-based accelerator with DES/3DES/AES/HMAC-SHA-1/HMAC-MD5 encryption, the Vigor3300 series supports low latency and high performance IPSec VPN for business application. Robust firewall features provide rich security services including Stateful Packet Inspection(SPI), access control, packet filtering, securing against intruders/attackers and Denial-of-Service (DoS). Integrated content filtering function can block inappropriate web contents for protection on campuses, or offices as well as monitor and block prohibited web contents for employees. The QoS (Quality-of-Service) function provides better services for selected network traffic over various technologies. The primary goal of QoS is to provide priority including dedicated bandwidth, controlled jitter and latency, and improve packet-loss characteristics. At the same time, these products ensure that providing priority for one or more flows does not make other flows fail. The load balance and high availability features improve the network performance, scalability, and reliability in Internet networks. View online demo click here
Svenska kronor4160.00

Intertex SurfinBird IX67 FW AIR GW Router
Modèle IX67AIR
Intertex
SIP capable firewall and router for the advanced user and offices up to 50 persons. 4 Ethernet ports (1 WAN and 3 LAN), 2 telephony ports (FXO and FXS) and 802.11b/g 54 Mbps wireless access point. - Easy installation and maintenance- Display and front panel keys- Auto login and keep-alive- Email notification- Dynamic DNS client- Quality of service- Network monitoring Adapter for IP Telephony (FXS) - Telephony is now offered over your broadband connection. The SurfinBird IX67 FW GW includes a port for connecting a conventional telephone for use with SIP based IP Telephony, without the use of extra adaptor box.Gateway to the Telephone Network (FXO) - The SurfinBird IX67 FW GW also includes a port to the telephone network, working as a gateway for SIP based IP telephony and the old telephone network. Use for emergency calls, IP telephony back-up and integration of your old telephony. Answer your old telephone and make local phone calls from your laptop when traveling.Support for SIP - SIP is the Internet standard for live person-to-person communication. The IX67 has a built-in SIP proxy and registrar that dynamically controls the firewall and NAT for full SIP support.The SIP Switch - With addition of the optional SIP Switch software, the IX67 becomes a complete telephony solution for the small office with unique functions and cost savings.VPN Server - There is olso an optional VPN termination software for the SurfinBird IX67 FW GW. This supports IPSec, PPTP and L2TP with certificate handling.
Svenska kronor2000.00

Intertex SurfinBird IX67 FW GW Router
Modèle IX67FWGW
Intertex
SIP capable firewall and router for the advanced user and offices up to 50 persons. 4 Ethernet ports (1 WAN and 3 LAN) and 2 telephony ports (FXO and FXS). - Easy installation and maintenance- Display and front panel keys- Auto login and keep-alive- Email notification- Dynamic DNS client- Quality of service- Network monitoringAdapter for IP Telephony (FXS) - Telephony is now offered over your broadband connection. The SurfinBird IX67 FW GW includes a port for connecting a conventional telephone for use with SIP based IP Telephony, without the use of extra adaptor box.Gateway to the Telephone Network (FXO) - The SurfinBird IX67 FW GW also includes a port to the telephone network, working as a gateway for SIP based IP telephony and the old telephone network. Use for emergency calls, IP telephony back-up and integration of your old telephony. Answer your old telephone and make local phone calls from your laptop when traveling.Support for SIP - SIP is the Internet standard for live person-to-person communication. The IX67 has a built-in SIP proxy and registrar that dynamically controls the firewall and NAT for full SIP support.The SIP Switch - With addition of the optional SIP Switch software, the IX67 becomes a complete telephony solution for the small office with unique functions and cost savings. VPN Server - There is olso an optional VPN termination software for the SurfinBird IX67 FW GW. This supports IPSec, PPTP and L2TP with certificate handling.
Svenska kronor1920.00

Linksys 24-port 10/100 + 2-Port GB Switch with POE
Modèle SRW224P
Linksys
The Linksys WebView Managed switch allows you to expand your network securely. Configuration of the switch is secured using SSL for Web access. User control is secured using 802.1x security using a RADIUS authentication mechanism and can also be controlled using MAC filtering. Extensive QoS features makes the solution ideal for real-time applications like Voice and Video. The 4 priority queues together with the Weighted Round Robin and Strict Priority scheduling techniques facilitate efficient co-existence of real-time traffic with data traffic allowing them each to meet their QoS needs. Individual users or applications can be prioritized above others using various Class of Service options - by port, layer 2 priority (802.1p), or Layer 3 priority (TOS or DSCP). Intelligent Broadcast and Multicast storm control minimizes and contain the effect of these types of traffic on regular traffic. IGMP Snooping limits bandwidth-intensive video traffic to only the requestors without flooding to all users. Incoming traffic can be policed allowing you to control network access and traffic flow. There are features that allow you to expand and grow your network of switches. Link aggregation allow multiple high-bandwidth trunks between switches to be setup. This also provides a level of reliability in that the system continues to operate if one of the links break. Spanning Tree(STP), Fast Spanning Tree, and Multiple Spanning Tree allow you to build a mesh of switches increasing the availibility of the system. The rich management functionality of the WebView switches includes SNMP, RMON, and HTTP Management options, allowing you to flexibly integrate and manage these devices in your network. Automatic Load Sensing - Power control circuitry automatically detects Power over Ethernet on access devices before providing power. Power feeding of Ethernet is provided on the 10/100 Base-T ports with maximum output power per port of up to 15.4W simultaneously on 12 ports. Alternatively, the switch can power all ports at a maximum of 7.5 W per port. Independent overload and short-circuit protection for each port is provided. LED indicators for power status are provided per port.
Svenska kronor3200.00

Linksys RT31P2 Voip Router
Modèle RT31P2
Linksys
Share your high-speed cable or DSL Internet connection with multiple computers- Built-in switch connects three local PCs directly -- daisy-chain out to more hubs and switches as your network grows- Two standard phone jacks enable feature-rich telephone service over your cable or DSL Internet connection- Supports Quality of Service for clear telephone reception, DHCP, and Universal Plug-and-Play (UPnP) Think of the Linksys Broadband Router with 2 Phone Ports as a kind of "splitter" for your Internet connection. Just connect your DSL or Cable Modem to the Router, and all the computers in your home or office can share the Internet -- all at the same time. The built-in 3-port switch lets you attach three local PCs directly, or daisy-chain out to more hubs and switches as your network grows. Once your computers are connected to the Internet through the Router, they can communicate with each other too, sharing resources and files. All your computers can print on a shared printer connected anywhere in the home. And you can share all kinds of files between computers -- music, digital pictures, and other documents. The two standard telephone jacks on the Router enable high-quality feature-rich telephone service through your high-speed connection even while you're surfing the Internet. Each phone jack operates independently -- like having two phone lines. With an appropriate Internet telephone service provider, you'll get low domestic and international phone rates, Caller ID, Call Waiting, Voicemail, Call Forwarding, Distinctive Ring, and lots of other available special phone features. Choose any free local dialing US area code, regardless of where you live. Add a virtual phone number in any area code, or even a US-wide toll-free number. It's all easier than you think -- the included Setup Wizard takes you through configuring the Router, step by step. The Router can act as a DHCP server for your network, so your PCs are configured automatically. Universal Plug-and-Play (UPnP) lets specialized Internet applications configure the Router so you don't have to. Built-in NAT technology helps keep intruders out of your computers. Advanced users can set up QoS prioritization based on port or service type. With the Linksys Broadband Router with 2 Phone Ports at the heart of your home network, you don't need to be a networking genius to share high-speed Internet, printers and files, and turn your Internet connection into a high-quality high-value telephone service.
Svenska kronor680.00

Linksys RTP300 Broadband Router with 2 Phone Ports
Modèle RTP300
Linksys
Think of the Linksys Broadband Router with 2 Phone Ports as a kind of "splitter" for your Internet connection. Just connect your DSL or Cable Modem to the Router, and all the computers in your home or office can share the Internet -- all at the same time. The built-in 4-port switch lets you attach four local PCs directly, or daisy-chain out to more hubs and switches as your network grows. Once your computers are connected to the Internet through the Router, they can communicate with each other too, sharing resources and files. All your computers can print on a shared printer connected anywhere in the home. And you can share all kinds of files between computers -- music, digital pictures, and other documents. The two standard telephone jacks on the Router enable high-quality feature-rich telephone service through your high-speed connection even while you're surfing the Internet. Each phone jack operates independently -- like having two phone lines. With Vonage Internet Phone Service, you'll get low domestic and international phone rates, and lots of other available special phone features. It's all easier than you think -- the included Setup Wizard takes you through configuring the Router, step by step. The Router can act as a DHCP server for your network, so your PCs are configured automatically. Universal Plug-and-Play (UPnP) lets specialized Internet applications configure the Router so you don't have to. The powerful SPI firewall helps keep intruders out of your computers. Advanced users can set up QoS prioritization based on port or service type. With the Linksys Broadband Router with 2 Phone Ports at the heart of your home network, you don't need to be a networking genius to share high-speed Internet, printers and files, and turn your Internet connection into a high-quality high-value telephone service.
Svenska kronor720.00

Linksys Wireless-G Broadband Router WRT54G
Modèle WRT54G
Belkin
Wireless-G is the upcoming 54Mbps wireless networking standard that's almost five times faster than the widely deployed Wireless-B (802.11b) products found in homes, businesses, and public wireless hotspots around the country - but since they share the same 2.4GHz radio band, Wireless-G devices can also interoperate with existing 11Mbps Wireless-B equipment.Since both standards are built in, you can protect your investment in existing 802.11b infrastructure, and migrate to the new screaming fast Wireless-G standard as your needs grow.The Linksys Wireless-G Broadband Router is really three devices in one box. First, there's the Wireless Access Point, which lets you connect Wireless-G or Wireless-B devices to the network. There's also a built-in 4-port full-duplex 10/100 Switch to connect your wired-Ethernet devices. Connect four PCs directly, or daisy-chain out to more hubs and switches to create as big a network as you need. Finally, the Router function ties it all together and lets your whole network share a high-speed cable or DSL Internet connection.To protect your data and privacy, the Wireless-G Broadband Router can encrypt all wireless transmissions. The Router can serve as a DHCP Server, has NAT technology to protect against Internet intruders, supports VPN pass-through, and can be configured to filter internal users' access to the Internet. Configuration is a snap with the web browser-based configuration utility.With the Linksys Wireless-G Broadband Router at the center of your home or office network, you can share a high-speed Internet connection, files, printers, and multi-player games with the flexibility, speed, and security you need!
Svenska kronor656.00

Linksys Wireless-G Broadband Router WRT54GP2
Modèle WRT54
Linksys
The Linksys Wireless-G Broadband Router is really four devices in one box. First, there's the Wireless Access Point, which lets you connect both screaming fast Wireless-G (802.11g at 54Mbps) and Wireless-B (802.11b at 11Mbps) devices to the network. There's also a built-in 3-port full-duplex 10/100 Switch to connect your wired-Ethernet devices together. Connect four PCs directly, or attach more hubs and switches to create as big a network as you need. The Router function lets your whole network share a high-speed cable or DSL Internet connection.The fourth function is the phone adapter which enables high-quality feature-rich telephone service through your high-speed connection even while you're surfing the Internet. There are two standard telephone jacks, each operating independently - like having two phone lines. With Vonage, you'll get low domestic and international phone rates, Caller ID, Call Waiting, Voicemail, Call Forwarding, Distinctive Ring, and lots of other available special phone features. Choose any free local dialing US area code, regardless of where you live. Add a virtual phone number in any area code, or even a US-wide toll-free number.To protect your data and privacy, the Wireless-G Broadband Router can encode all wireless transmissions with up to 256-bit encryption, and supports both Wired Equivalent Privacy (WEP) and the industrial-strength wireless security of Wi-Fi Protected Access (WPA). The Router can serve as a DHCP Server, has a powerful SPI firewall to protect your PCs against intruders and most known Internet attacks, supports VPN pass-through, and can be configured to filter internal users' access to the Internet. Configuration is a snap with the web browser-based configuration utility.With the Linksys Wireless-G Broadband Router at the center of your home or office network, you can share a high-speed Internet connection, files, printers, and multi-player games, and turn that Internet connection into a high-quality, high-value telephone service!
Svenska kronor1040.00

Linksys WRT54GC router
Modèle WRT54GC
Linksys
- All-in-one Internet-sharing Router, 4-port Switch, and Wireless-G (802.11g) Access Point- Shares a single Internet connection and other resources with Ethernet wired and Wireless-G and B devices- Compact form factor fits in anywhere- High security: Wi-Fi Protected Access™ (WPA/WPA2 Personal), wireless MAC address filtering, powerful SPI firewall The Linksys Compact Wireless-G Broadband Router is really three devices in one small box. First, there's the Wireless Access Point, which lets you connect both screaming fast Wireless-G (802.11g at 54Mbps) and Wireless-B (802.11b at 11Mbps) devices to the network. There's also a built-in 4-port full-duplex 10/100 Switch to connect your wired-Ethernet devices together. Connect four PCs directly, or attach more hubs and switches to create as big a network as you need. Finally, the Router function ties it all together and lets your whole network share a high-speed cable or DSL Internet connection. Once your computers are connected to the Router and the Internet, they can communicate with each other too, sharing a printer, digital music, picture and document files, and multi-player or on-line games. To protect your data and privacy, the Wireless-G Broadband Router can encode all wireless transmissions with WEP or industrial-strength WPA Personal encryption. The Router can serve as a DHCP Server, has a powerful SPI firewall to protect your PCs against intruders and most known Internet attacks, and supports VPN pass-through. Configuration is a snap with the web browser-based configuration utility. And the tiny form factor fits any home or desktop. With the Linksys Compact Wireless-G Broadband Router at the center of your home or office network, you can share a high-speed Internet connection, files, printers, and multi-player games with the flexibility, speed, and security you need!
Svenska kronor496.00

Linksys WRTP54G VoIP Wireless Router
Modèle WRTP54G
Linksys
The Linksys Wireless-G Broadband Router is really four devices in one box. First, there's the Wireless Access Point, which lets you connect both screaming fast Wireless-G (802.11g at 54Mbps) and Wireless-B (802.11b at 11Mbps) devices to the network. There's also a built-in 4-port full-duplex 10/100 Switch to connect your wired-Ethernet devices together. Connect four PCs directly, or attach more hubs and switches to create as big a network as you need. The Router function lets your whole network share a high-speed cable or DSL Internet connection. The fourth function is the phone adapter which enables high-quality feature-rich telephone service through your high-speed connection even while you're surfing the Internet. There are two standard telephone jacks, each operating independently -- like having two phone lines. With Vonage Service, you'll get low domestic and international phone rates, Caller ID, Call Waiting, Voicemail, Call Forwarding, Distinctive Ring, and lots of other available special phone features. Choose any free local dialing US area code, regardless of where you live. Add a virtual phone number in any area code, or even a US-wide toll-free number. To protect your data and privacy, the Wireless-G Broadband Router can encode all wireless transmissions with up to 128-bit encryption, and supports both Wired Equivalent Privacy (WEP) and the industrial-strength wireless security of Wi-Fi Protected Access™ (WPA). The Router can serve as a DHCP Server, has a powerful SPI firewall to protect your PCs against intruders and most known Internet attacks, supports VPN pass-through, and can be configured to filter internal users' access to the Internet. Configuration is a snap with the web browser-based configuration utility. With the Linksys Wireless-G Broadband Router at the center of your home or office network, you can share a high-speed Internet connection, files, printers, and multi-player games, and turn that Internet connection into a high-quality, high-value telephone service!
~~Svenska kronor880.00~~ **Svenska kronor840.00**

Voip Gateways

Modèle VOI-4100
Level One
- 4 x FXS RJ-11 voice ports- 2 x10/100 Mbps RJ-45 ports: 1xWAN; 1xLAN- Supports SIP v2.0 (RFC3261) voice protocol- Supports G.723.1 (6.3k/ 5.3k), G.711(A-law/£g-law) and G.729A/B Voice Codec- Supports G.168 Echo cancellation- Supports DTMF detection/Generation- Supports CNG, VAD voice processing- Built-in HTTP server for setting through web browser- Remote software upgrade through TFTP
Svenska kronor2720.00

Grandstream GXW-4004 FXS Analogue Gateway
Modèle GXW4004
Granstream Networks
The Grandstream GXW-4004 offers 4 RJ-11 FXO ports for connecting and managing analogue devices. FXS gateways allow to convert outgoing PSTN calls to VoIP calls. The configuration is easily effected through a clear web interface. With its PSTN failover function it also works while power failure.Main features- 4x FXS analogue port - converts outgoing PSTN calls to VoIP calls- PSTN failoverTechnical specifications - Standards: IP, TCP, DHCP, TFTP, HTTP, HTTPS, RTP/RTCP - Connection Port: 2x RJ-45 10/100T- Base Ethernet port,   4x RJ-11 FXS port- Protocol: SIP - Audio CODEC: G.711, G.729A/B/E, G.723, G.726, G.728, iLBC - Management: HTTP/HTTPS web interface - Fax support: T.38 - Dimensions: 135 mm x 230 mm x 35 mm (WxDxH)
Svenska kronor1400.00

Grandstream GXW-4008 FXS Analogue Gateway
Modèle GXW4008
Granstream Networks
The Grandstream GXW-4008 offers 8 RJ-11 FXO ports for connecting and managing analogue devices. FXS gateways allow to convert outgoing PSTN calls to VoIP calls. The configuration is easily effected through a clear web interface. With its PSTN failover function it also works while power failure.Main features- 8x FXS analogue port - converts outgoing PSTN calls to VoIP calls- PSTN failoverTechnical specifications - Standards: IP, TCP, DHCP, TFTP, HTTP, HTTPS, RTP/RTCP - Connection Port: 2x RJ-45 10/100T- Base Ethernet port,   8x RJ-11 FXS port- Protocol: SIP - Audio CODEC: G.711, G.729A/B/E, G.723, G.726, G.728, iLBC - Management: HTTP/HTTPS web interface - Fax support: T.38 - Dimensions: 135 mm x 230 mm x 35 mm (WxDxH)
Svenska kronor1920.00

Grandstream GXW-4104 FXS Analogue Gateway
Modèle GXW4104
Granstream Networks
The Grandstream GXW-4104 offers 4 RJ-11 FXO ports for connecting and managing analogue devices. FXO gateways allow to convert incoming PSTN calls to VoIP calls. The configuration is easily effected through a clear web interface. The gateway provides 3 SIP- lines/profiles per system. Its build- in video surveillance port allows functions like security monitoring. Main features- 4x FXO analogue port - 3 SIP- lines/profiles per system- Video surveillance port Technical specifications - Standards: IP, TCP, DHCP, PPPoE, NAT, RTP/RTCP - Connection Port: 2x RJ-45 10/100T- Base Ethernet port,   4x RJ-11 FXO port, video surveillance port - Protocol: SIP - Audio CODEC: G.711, G.729A/B, G.723, GSM, G.726, H.264 - Management: HTTP/HTTPS web interface - Fax support: T.38 - Dimensions: 172 mm x 225 mm x 42 mm (WxDxH)
Svenska kronor2200.00

Grandstream GXW-4108 FXS Analogue Gateway
Modèle GXW4108
Granstream Networks
The Grandstream GXW-4108 offers 8 RJ-11 FXO ports for connecting and managing analogue devices. FXO gateways allow to convert incoming PSTN calls to VoIP calls. The configuration is easily effected through a clear web interface. The gateway provides 3 SIP- lines/profiles per system. Its build- in video surveillance port allows functions like security monitoring.Main features- 8x FXO analogue port - 3 SIP- lines/profiles per system- Video surveillance port Technical specifications - Standards: IP, TCP, DHCP, PPPoE, NAT, RTP/RTCP - Connection Port: 2x RJ-45 10/100T- Base Ethernet port,   8x RJ-11 FXO port, video surveillance port - Protocol: SIP - Audio CODEC: G.711, G.729A/B, G.723, GSM, G.726, H.264 - Management: HTTP/HTTPS web interface - Fax support: T.38 - Dimensions: 172 mm x 225 mm x 42 mm (WxDxH)
Svenska kronor3040.00

Lancom 1722 Voip
Modèle 1722V
Lancom Systems
VoIP PBX and switching functions for up to 32 SIP subscribers SIP proxy for registration with providers and upstream VoIP PBXs SIP gateway with transparent transition between SIP and ISDN Intelligent call routing and number translation Two ISDN ports (NT/TE) with Life-Line Support Stateful-inspection firewall and VPN gateway LANCOM 1722 VoIP is the innovative all-round solution for integrated data and voice networking for interconnecting company sites. It provides a PBX, VPN router, access router and professional firewall in one compact device. The business VoIP gateway LANCOM 1722 VoIP seamlessly switches between ISDN telephones, existing ISDN PBXs, SIP subscribers and central VoIP PBXs. By enabling the continued use of existing ISDN technology, the LANCOM 1722 VoIP offers investment protection and cost savings with Voice over IP for small and mid-sized enterprises, home offices and subsidiaries. More Telephony. Voice connections can be automatically directed over existing VPN connections between company sites. Not only are these calls secured from interception; by using flatrate connections, there are no additional costs, either. Small and mid-sized companies can even save themselves complete local PBXs by equipping new installations with a LANCOM VoIP router instead. LANCOM is opening up a cost-effective migration path between technologies of the “old and new world”—existing ISDN telephones and PBXs can be integrated and operated side by side with VoIP terminal equipment. Two ISDN ports can be freely configured as internal or external connections and offer four parallel ISDN voice channels. This allows, for example, an existing ISDN PBX to be additionally equipped with SIP and connected to an upstream VoIP PBX. Subscribers can simultaneously make calls via ISDN telephones, SIP equipment, or softphones to other SIP or ISDN subscribers, both internally and externally. The transition between SIP and ISDN is automatic and invisible to the user. With Life-Line Support and power relay to the internal ISDN port, it is still possible to telephone with ISDN even in case of a power outage. ISDN backup, load balancing and VRRP combined with up to three Ethernet ports as WAN interfaces provide SIP connections with redundancy and high reliability. This ensures that telephony is just as reliable as ever, even with VoIP.
Svenska kronor4520.00

Lancom 1724 Voip
Modèle 1724V
Lancom Systems
Four ISDN ports (NT/TE) for 8 parallel ISDN channels VoIP switching functions for up to 32 SIP subscribers SIP proxy for registration with providers and upstream VoIP PBXs SIP/ISDN gateway with transparent transition between SIP and ISDN Intelligent call routing and number translation Stateful-inspection firewall and VPN gateway   The LANCOM 1724 VoIP is ideally suited to users requiring multiple So buses to translate ISDN to SIP and vice versa. In a single compact device, this VoIP gateway features a VPN router, access router and professional firewall for a powerful integration of voice and data services in day-to-day business. The business VoIP gateway LANCOM 1724 VoIP seamlessly switches between ISDN telephones, existing ISDN PBXs, SIP subscribers and central VoIP PBXs. Four switchable ISDN ports for ISDN point-to-multipoint connections and for ISDN point-to-point connections provide flexibility and protection of investment for the migration of existing ISDN PBXs to Voice over IP. This device is ideal for small and mid-size locations with up to 32 direct SIP telephony subscribers and transition of any number of telephony subscribers from ISDN to SIP servers.   More Telephony. Voice connections can be automatically directed over existing VPN connections between company sites. Not only are these calls secured from interception; by using flatrate connections, there are no additional costs, either. Small and mid-sized companies can even save themselves complete local PBXs by equipping new installations with a LANCOM VoIP router instead. LANCOM opens up a cost-effective migration path between technologies of the "old and new world" because existing ISDN telephones and PBXs can be integrated and operated side by side with VoIP terminal equipment. Four ISDN ports can be freely configured as internal or external connections and offer eight parallel ISDN voice channels. This allows, for example, an existing ISDN PBX to be additionally equipped with SIP and connected to an upstream VoIP PBX. Subscribers can simultaneously make calls via ISDN telephones, SIP equipment, or softphones to other SIP or ISDN subscribers, both internally and externally. The transition between SIP and ISDN is automatic and invisible to the user. ISDN backup, load balancing and VRRP combined with an Ethernet port as an additional WAN interface provide SIP connections with redundancy and high reliability. This ensures that telephony is just as reliable as ever, even with VoIP.
Svenska kronor5600.00

IP-PBX

Linksys SPA-9000 IP-PBX
Modèle SPA9000
Linksys
The SPA9000 marries the rich feature set of high-end PBX telephone systems with the convenience and cost advantages of Voice over IP. It has common voice system features such as an auto-attendant, shared line appearances, three way call conferencing, intercom, music on hold, call-forwarding and much more. The SPA9000 opens up access to the benefits of VoIP, including low cost long distance service, telephone number portability, and one network for both voice and data. The SPA9000 is so easy to configure that a fully working system can be set up in minutes. New telephones are automatically detected and registered when they are connected to the SPA9000. The SPA9000 has an integrated web server that allow features to be configured using a web browser. The web server has multiple levels of password protected access to user and service level features. Service level settings may be locked by the Internet Telephone Service Provider to ensure they are not inadvertently corrupted. The Internet Telephone Service Provider also can remotely update the software and settings through a secure encrypted connection. With its integrated router, the SPA9000 can be either connected directly to the internet connection or to another router on your network. The SPA9000 has separate WAN and LAN Ethernet ports. The WAN connection can be connect through DHCP or a fixed IP address. The LAN port can assign IP addresses to IP Phones and computers using NAT and DHCP. While the SPA9000 will work with any SIP compatible IP Phone, it is the ideal host for Linksys IP Phones, such as the SPA901, SPA921, SPA922, SPA941, and SPA942. Powerful configuration capabilities enable the SPA9000 to support a greater set of advanced features with these IP Phones, such as shared line appearances, hunt groups, call transfer, call parking lot, and group paging. With its two FXS ports, the SPA9000 can support traditional analog devices such as telephones, answering machines, FAX machines, and media adapters. Watch a Demonstration There are special benefits that make the Linksys Voice System 9000 easy to install and use. See the demonstration on how the Linksys Voice System 9000 can work for your business. Flash Demonstration
~~**Svenska kronor2560.00**~~ **Svenska kronor2160.00**

MeshVPN

Fiberlogic MeshVPN 2320
Modèle VPN2320
Fiberlogic
The MeshVPN 2320, with its state of the art design, provides a complete Package for VPN failover, reliability, redundancy, speed and security for transmissions by load balancing multiple VPN connections. The MeshVPN 2320 series can aggregate up to two WAN links which yield two times bandwidth with inbound/outbound traffic management to achieve outstanding VPN trunking and VPN load balancing between them. Besides from from the features mentioned above, MeshVPN 2300 series also consider security as an important issue. The MeshVPN 2300 series guard your system with SPI firewall, anti DoS, IDS/IPS and more. Higher Standard VPN Connection: Up to 70 site-to-site VPN tunnels Up to 300 Road Warrior VPN tunnels IPSec NAT traversal IPSec / L2TP / PPTP MD5 / SHA1 authentication algorithm Up to 35 Mbps VPN throughput with 3DES encryption protection Faster Computation and Higher Hardware Demanding: Up to PowerPC 266 MHz Up to 64 MB SDRAM Up to 8 MB Flash Up to 2 WAN Ports to different ISPs (either ADSL, Cable, or mixed) Perfectly Efficient Bandwidth Usage: Up to 50 Mbps load balanced throughput Up to 50000 concurrent sessions Up to 35000 new sessions per second Higher Security: DMZ host anti DoS NAT / PAT / rule based NAT IDS / IPS High Availability: Device failure detection Failover notification Friendly Management: Remote login Remote firmware upgrade Quick install - SNMP / MRTG / SSH v1,v2

Svenska kronor12800.00

Fiberlogic MeshVPN 2340
Modèle VPN2340
Fiberlogic
The MeshVPN 2340, with its state of the art design, provides a complete package for VPN failover, reliability, redundancy, speed and security for transmissions by load balancing multiple VPN connections. It can aggregate up to 4 WAN links which yield for times bandwidth with inbound /outbound traffic management to achieve outstanding VPN trunking and VPN load balancing between them. Besides from the features mentioned above, MeshVPN 2300 series also consider security as an important issue. Higher Standard VPN Connection: Up to 150 site-to-site VPN tunnels Up to 500 Road Warrior VPN tunnels IPSec NAT traversal IPSec / L2TP / PPTP MD5 / SHA1 authentification algorithm Up to 60 Mbps VPN throughput with 3DES encryption protection Faster Computation and Higher Hardware Demanding: Up to PowerPC 266 MHz Up to 128 MB SDRAM Up to 16 MB Flash Up to 4 WAN ports to different ISPs (either ADSL, cable, or mixed) Perfectly Efficient Bandwidth Usage: Up to 75 Mbps load balance throughput Up to 100000 concurrent sessions Up to 60000 new sessions per second Higher Security: DMZ host anti DoS NAT / PAT / rule based NAT IDS / IPS High Availability: Virtual Server HA VRRP - hardware error detection Failover notification Friendly Management: Remote login Remote firmware upgrade Quick install SNMP / MRTG / SSH v1,v2
Svenska kronor58000.00

Fiberlogic MeshVPN 2380
Modèle VPN2380
Linksys
The MeshVPN 2300 Series, with its state of the art design, provides a complete package for VPN failover, reliability, redundancy, speed and security for transmissions by load balancing multiple VPN connections. The MeshVPN 2300 series can aggregate up to eight WAN links which yield eight times the bandwidth with inbound/outbound traffic management to achieve outstanding VPN-trunking and VPN load balancing between them. Besides from the features mentioned above, MeshVPN 2300 series also consider security as a very important issue. The MeshVPN 2300 series guard your system with SPI firewall, Anti-DoS, IDS/IPS and more.   Ideal for businesses which require: HyperDNS Destination NAT Mesh VPN: Allows you to connect mutiple branch office areas as your own local LAN. VRRP: Provides fault tolerant/backup capability to the network in case one unit might be down. The backup unit will take over and become the master one and still service the whole network. Higher Standard VPN Connection: Up to 300 site-to-site VPN tunnels Up to 800 Road Warrior VPN tunnels IPSec NAT traversal IPSec / L2TP / PPTP MD5 / SHA1 authentication algorithm Up to 100 Mbps VPN throughput with 3DES encryption protection Faster Computation and Higher Hardware Demanding: Up to PowerPC 333 MHz Up to 192 MB SDRAM Up to 16 MB Flash Up to 8 WAN ports to different ISPs (either ADSL, cable, or mixed) Perfectly Efficient Bandwidth Usage: Up to 100 Mbps load balance throughput Up to 20.0000 concurrent sessions Up to 10.0000 new sessions per second Higher Security: SPI firewall (stateful packet inspection) NAT / PAT / rule based NAT ANTI-DoS / IDS / IPS High Availability: Virtual Server HA VRRP - error detection Failover notification Friendly Management:

Remote login
Remote firmware upgrade
Quick install
SNMP / MRTG / SSH v1,v2
Svenska kronor125600.00

DECT with VoIP

Siemens 675 IP telephone
Modèle S675
Siemens
The fashionable phone for intuitive elegance plus an integrated answering machine If you're looking for a full-featured phone that combines fashionable design with intuitive operation and which also features the convenience of an integrated answering machine, take a closer look at the award-winning* Siemens Gigaset S675. Featuring a big, brilliant color display and intuitive icon-based menu – this phone is both extremely easy on the eyes and a joy to use. An easy navigation key lets you effortlessly access all of the phone's key features and makes operation a truly intuitive experience. It even comes with programmable soft keys for putting the features you use most often right at your fingertips. Designed for people who demand more from their home phone, the high-tech Siemens Gigaset S675 also offers you an array of advanced features that are the perfect complement to its fashionable design. Integrated answering machine You'll never miss another important call thanks to this phone's integrated answering machine featuring up to 40 minutes of recording time. New messages are displayed on the phone's handset, so you'll know someone tried to reach you even if you're away from the base station. You can also easily keep track of who called and when because the Siemens Gigaset S675 announces the date and time of new messages by voice. For added convenience you can operate the answering machine via the handset or base station, so you can check up on new messages from wherever you are in your house. If you're away from home this phone will even alert you when you receive a new message via SMS**. Effortlessly manage all your contacts Always have the names and numbers you need right at your fingertips. The Siemens Gigaset S675 gives you a phonebook to match its fashionable design – offering you up to 250 vCard entries for managing your social life. Store the first and last name of each of your contacts as well as 3 numbers, e-mail address and birthday. Thanks to this phone's big color display and icon-based menu, accessing all your vCard contacts is effortlessly intuitive. SMS your friends and family Enjoy all the added convenience of SMS** texting. The Siemens Gigaset S675 lets you send messages with up to 640 characters and even send and receive e-mails via SMS. Experience advanced handsfree calling Designed to give you more calling freedom, the fully-featured Siemens Gigaset S675 lets you make handsfree phone calls in brilliant sound quality with the convenience of adjustable volume control. That means you're free to chat with your friends while you get on with other things around the house. Save energy with ECO DECT Featuring an energy-saving power supply, the Siemens Gigaset S675 uses up to 60% less energy than conventional phones – so it saves you money and is better for the environment. Like all Gigaset cordless phones, this model variably reduces transmitting power from the handset to base station according to their distance apart. Thanks to integrated ECO DECT technology, it also goes further. Choosing ECO Mode allows you to reduce the base station's transmitting power by 80%. When docked***, the transmitting power of both handset and base station are also automatically reduced to a minimum. As a result, you can enjoy a more energy-efficient way of staying connected with the world. Compatible with the HC450 door intercom Welcome your guests the easy way. Together with the Siemens Gigaset HC450 door intercom this phone can be transformed into a remote control for your front door. For more info rmation on this feature, head to easy-welcome.com. Enjoy the pleasure of communicating. The Siemens Gigaset S675 is a joy to operate and offers you the best of advanced features and award-winning fashionable design – plus an integrated answering machine . Phoning has never been more intuitive or elegant.
Svenska kronor1120.00

Siemens C450 IP cordless phone
Modèle C450IP
Siemens
With the Gigaset C450 IP, you don’t even need a PC to start making internet phone calls. If you like flexibility, then you’ll love how easy it is to switch between internet and fixed-line calls. Just press a button. It’s that easy, especially thanks to plug-and-play convenience too.The Gigaset C450 IP proves that cutting-edge technology can also be brilliantly easy to use. Without the need for a PC, you simply plug-in the Gigaset C450 IP to your phone connection and in a few steps you can start phoning over the internet to your friends and family around the world. Conveniently select the internet telephony provider you prefer on the handset’s illuminated display. No matter what, you’re assured Gigaset’s Quality of Service (QoS) for superb internet telephony quality. VoIP doesn’t get any easier than that. Highlights:- Dual mode: Easy switch from VoIP calls to fixed line calls by single keypress - Easy configuration of internet telephony (VoIP) without a PC - 1 VoIP call and 1 fixed line call in parallel with multiple handsets - Convenient handsfree talking Features:- Hybrid: IP and fixed-net telephony- Easy set-up of IP telephony with or without a PC- Illuminated color display and keypad- Phonebook for up to 100 entries- Provider setting directly at the handset- Ethernet (IEEE 802.3) and PSTN access- Plug & Play: the unit can be used without special installation (unpack the phone, connect the phone and conduct a call)- Internal antennas- Date and time functions protected against power failure- Protocol support: SIP, RTP,DHCP,STUN- Size handset: 141x53x31mm (LxWxH)- GAP compatible- Battery: Talktime ~13h, Standby ~125h- Codecs: g711,g726,g729A/B, VAD/CNG- QoS: TOS, DiffServ View flash presentation klick here
Svenska kronor780.00

Siemens C470IP DECT telephone
Modèle C470IP
Siemens
Dual mode: easy switch from internet calls to fixed-line calls by single keypress. Information services as screensaver* e.g. weather forecast Expandable phone system with multiline calling for up to 6 handsets and 6 SIP accounts from different providers ECO DECT – for 60% less energy use[1] and variable reduction of transmitting power   The attractive VoIP and fixed-line phone for up to 3 parallel calls Take part in the VoIP revolution and benefit from cost-efficient internet telephony with the attractive Gigaset C470 IP cordless phone. Specially designed to match your home’s sophisticated décor, this smart phone features dual mode functionality, enabling you to easily switch between internet and fixed-line calls at the touch of a button. For ultimate flexibility, this model lets you make internet calls without a PC and access up to 6 SIP accounts from different VoIP providers directly on the handset. It’s also an expandable phone system that will support up to 6 handsets, giving you the freedom to make up to 3 parallel calls . If you’re waiting for an important e-mail message, the Gigaset C470 IP even offers a handy e-mail notification feature, which alerts you of any new messages on the handset – including the date, time and subject line of the mails. This Gigaset model also features new energy-saving ECO DECT technology, which uses up to 60% less energy than our conventional cordless phones – so it saves you money and is friendlier to the environment.   If you’re looking for an attractive dual-mode phone that gives you the freedom to make VoIP and fixed-line calls, then look no further than the Gigaset C470 IP. Dual mode and parallel calling convenience

To give you maximum flexibility, the Gigaset C470 IP lets you easily switch between internet and fixed-line phoning at the touch of a button – whichever suits you best. You can also add up to 6 extra handsets to your phone system and experience the convenience of multiline calling . That means up to 3 family members or friends can make parallel calls – up to 2 VoIP calls and 1 fixed-line call at the same time.
Advanced messaging options

The Gigaset C470 IP offers you a host of extra options for staying in touch with family and friends around the world. The handy e-mail notification function sends you a message on the handset's illuminated display with the time, date and subject of any new e-mails you receive – without you having to be online. If you're away from home, you can also program the Gigaset C470 IP to alert you of any missed calls by sending you an SMS[2] to your cellphone or another external number. Naturally, this phone lets you send SMS messages too – perfect for those times when you'd love to chat, but just don't have time.

Convenient handsfree talking in brilliant sound quality
This phone's handsfree function is a great solution for multi-taskers with busy lifestyles. It lets you carry on with the day's tasks or just lie back on the couch with your hands behind your head while you enjoy a relaxing chat in brilliant sound quality.
A phonebook for easily managing busy social lives
Do you have loads of contacts and want to conveniently store them in one place? The Gigaset C470 IP features an extra-large phonebook that lets you manage the numbers of up to 150 of your family members and friends. Finding, editing or deleting your contacts is also simple thanks to the easy-to-read illuminated display and intuitive icon-based menu.

Your world's at your fingertips with Gigaset.net
There's no quicker way to start phoning with your loved ones over the internet than with Gigaset.net . Purchase the Gigaset C470 IP phone and you'll automatically become a member of this exclusive Gigaset IP network community. That means you'll be able to make VoIP calls to anyone else with a Gigaset IP phone – anywhere and anytime for free – and you won't even need an account with a VoIP provider. What's more, you'll be able to access free subscription services such as RSS feeds and weather information directly via your Gigaset handset.
ECO DECT for saving energy and costs

Thanks to an energy-saving power supply, the Gigaset C470 IP uses up to 60% less energy than our conventional cordless phones – making it better for the environment and more cost efficient. Like all Gigaset cordless phones, it also variably reduces transmitting power from the handset to base station according to their distance apart – but goes further in a number of other ways. Choosing ECO Mode on this model reduces the base station's transmission power by 80%[3]. As a result, you can enjoy a more energy-efficient way of staying connected with your world.
An attractive dual-mode phone specially designed for keeping in touch with friends and family around the world, the Gigaset C470 IP offers you all the convenience of cordless VoIP and fixed-line telephony.

Easy configuration of internet telephony (VoIP) without a PC
Gigaset.net: Worldwide free of charge calls between Gigaset.net supporting VoIP phones
Up to 3 calls in parallel: 2 VoIP calls and 1 fixed-line call with multiple handsets
E-mail notification with time, date and subject

Instant Messaging (buddy list, chatting, presence status)
Increased virus protection thanks to protected operating system
Convenient handsfree talking in brilliant sound quality
CLIP / CNIP
Illuminated display with 4096 colors and 128 x 128 pixels
Intuitive icon-based menu
SMS with up to 612 characters
Phonebook for up to 150 names and numbers
Svenska kronor880.00

Siemens Gigaset S45 DECT telephone
Modèle S45
Siemens
The Gigaset S45 features the ultimate in convenience: there is no shortage of features, but everything remains easy to operate. The Gigaset S45 combines modern communication and highest-level comfort with pure simplicity. Speed dialing, directory transfer and a fully duplex handsfree function – whatever the feature, they all meet the very highest standards of classic convenience.   When you expand your Gigaset basestation with more handsets, such as the Gigaset S45, you can place them in several rooms of your apartment – the end of having to run around the house to get a call.
Svenska kronor640.00

Siemens S450 IP cordless phone
Modèle S450IP
Siemens
The Gigaset S450 IP puts everything at your fingertips. Switch between making internet calls and fixed-line calls - no PC required! In addition to plug-and-play simplicity, this VoIP-ready phone tells you if your friends are online and even notifies you if you’ve received any e-mails. The Gigaset S450 IP proves that cutting-edge technology can also be brilliantly easy to use. Without the need for a PC, you simply plug-in the Gigaset S450 IP to your phone connection and in a few steps you can start phoning over the internet to your friends and family around the world. Conveniently select the internet telephony provider you prefer on the handset’s illuminated display. No matter what, you’re assured Gigaset’s Quality of Service (QoS) for superb internet telephony quality. VoIP doesn’t get any easier than that.- Dual mode: Easy switch from VoIP calls to fixed line calls by single keypress - 2 internet calls in parallel or 1 fixed-line call and 1 VoIP call in parallel with multiple handsets - Multiline: assign up to 4 VoIP numbers* for multihandset use - E-mail notification with time, date and subject* *Dependent on VoIP provider
Svenska kronor1040.00

Snom M3 DECT IP telephone
Modèle M3
Snom
Moving snom phones away from the desktop into cordless territory, the snom m3 is snom's first IP DECT phone offering – providing the latest in elegance and mobility without compromising audio quality.   At a Glance   Display: 128 x 128 pixels, 65536 colors, backlit Li-Ion battery pack for 8 hours of calls or 100 hours standby Range: 50 meters indoors, 100 meters outdoors 12 numerical keys, 5 navigation keys, 2 function keys Speakerphone on mobile handset Polyphonic ringtones Automatic registration of handset Separate charging cradle for handset 8 handsets per base station 8 SIP registrations with different servers/registrars Up to 3 concurrent calls per base station Three-way conference Remote setup, password protection Open DECT GAP standard
Svenska kronor1600.00

WiFi Radio

TerraTec Noxon iRadio
Modèle N100
TerraTec
The future of radioThe NOXON iRadio gives you free access to thousands of Internet radio stations worldwide. Chill to relaxed electronica from the world's most mellow DJs. Experience the latest in US college radio, Chinese dance tracks or Swedish metal. Get fresh information from home anywhere in the world, or delve into music and talk radio from faraway lands. Whatever you do, the compact radio design of the NOXON iRadio makes new listening opportunities as uncomplicated as possible. Switch on the NOXON iRadio and a wizard integrates it into your existing network. Configure a few settings, and the full range of Internet radio is at your fingertips. The stations are sorted according to countries and genres to help you keep an overview while exploring. The NOXON iRadio also makes the music on the hard drive of your computer available anywhere in the house. Simply install the included software for Windows, Mac or Linux to read all of the required title information from the MP3 and WMA files on your computer. The neatly-sorted songs will be available immediately for wireless listening! The NOXON iRadio is simple and intuitive to use. All of its functions can be controlled either from the included remote, or using buttons directly on the device. Its integrated display provides all the information you need. The integrated speaker delivers solid bass and clean treble. Package Content: Noxon iRadio Power adapter Remote control (with batteries) Software CD-ROM Manual Registration card Service card
Svenska kronor1840.00

ISDN cards

JUNGHANNS.NET quadBRI 2.0 ISDN card
Modèle JBRI2
Junghanns.net
Multiple quadBRI 2.0 PCI ISDN cards can be interconnected over an external PCM bus. The card's active channel switching capability (to bridge B channels on the card without latency and not using the host CPU) is also working over the external PCM bus (e.g. B channels on different cards can be actively switched). Target applications ISDN BRI PBX ISDN least cost routers Voice over IP BRI gateways VoIP integration of ISDN equipment PBX to PBX VoIP trunkingRequirements CPU 500+ Mhz RAM 64+ MB Linux 2.4.X or 2.6.X Kernel (2.6.X recommended) 32bit PCI Version 2.2 slot Features 4 Basic Rate Interface ports (I.421) for TE and NT mode DTMF detection Conference bridge PCM bus connectors daisy chaining of max. 8 cards 4 dual-color LEDs (layer 1 state indicators) active channel switching (across multiple cards over the external PCM bus) Point-to-Point (TE / NT) and Point-to-Multipoint(TE / NT) euroISDN protocol stack suitable for 3.3 volts and 5.0 volts 32 bit PCI slots hardware watchdog optional onboard S0 bus power feeding module available
Svenska kronor4400.00

Headsets

GN NETCOM 9120 MidiBoom GAP headset
Modèle GN9120
GN Netcom
Truly wireless headset for complete freedom and ease of use. Full mobility in the office ? up to 150m range & up to a 12-hour talk time. Unique CC Benefit: allows up to 4 agents to be trained at once Unique CC Benefit: can operate 2-3 headsets from one base unit and recharge on additional units Unique CC Benefit: wide range of optional extras ? choice of 3 boom arms, noise cancelling, filtering and standard microphones & remote call answering / ending Ultra-lightweight: weighs the same as a traditional-style headset Telephone conferencing: fully mobile, hassle-free conference calling on the move only for GAP-compatible DECT-Telephones The GN 9120 is a wireless headset that can revolutionise the way staff operate in the workplace be it office or contact centre. Unparalleled personal mobility allows workers to move freely about the facility to refer to files, reference points or products whilst on the phone, dealing more efficiently and quickly with the customer. With unbeatable acoustic performance over a wide range, a robust talk time and a host of innovative features and benefits, the revolutionary GN 9120 will transform your workplace. The unique conference call feature lets you hold fully secure conference calls on the move without being tiedto meeting rooms or ruined by crackling speakerphones ? also particularly useful for training purposes in contact centres where up to four agents can listen-in to a call. The GN 9120 comes with a host of accessory options like remote call answering/ending, which lets you answer/end calls while away from your desk. And, you can even buy additional headsets for colleagues or visitors or an extra base unit for charging spare headsets ? ideal for shift patterns in contact centres. Naturally, it?s compliant with the toughest acoustic shock protection standards, and is compatible with virtually any desk telephone.
Svenska kronor1200.00

GN Netcom 9330 headset
Modèle GN9330
GN Netcom
Det nya trådlösa headsetet GN 9330 USB kommer att få alla dina IP-telefonisamtal via datorn att låta bättre. Med headsetets revolutionerande ljud med frekvensomfång Wideband får ljudet en extra dimension. Ljudet blir tydligare och mer detaljrikt. Aldrig tidigare har det varit så här enkelt att installera och använda ett headset. Det går mycket snabbt att ansluta GN 9330 USB till datorn. Stylish, lightweight design Excellent sound clarity Up to 9 hours talk time Up to 100 m. of wireless freedom Multiple wearing styles included GN 9330 is designed to work with traditional desk telephones bringing hands-free convenience and superior audio quality to conventional office telephony. GN 9330 provides you with clear, peak-free sound. Noise-cancelling microphone technology ensures your voice is always transmitted clearly, even when you?re moving through noisy areas. The GN 9330 headset connects effortlessly and quickly to your telephone. Simple, intuitive headset controls make it easy to operate and with a weight of only 26 grams you?ll forget you are even wearing it. With a range of up to 100 metres, you can take the call all over the office environment, reaping the benefits of hands-free flexibility. If you want to escape the limitations set by your traditional telephone handset, the GN 9330 high-performance wireless headset pushes the boundaries, providing optimum range, all-day comfort and sound clarity.
Svenska kronor2000.00

Plantronics CS351 headset
Modèle PCS351
Plantronics
Office professionals with phone-intensive responsibilities can now answer or end calls while moving about the office?in unmatched comfort and with premium audio quality. Built on digitally-encrypted DECT 6.0 technology, the Plantronics SupraPlus® Wireless Professional Headset System combines the clarity of traditional corded technology with the mobility of wireless, so conversations can continue up to 300 feet from a desk phone. Product Features: Noise-Canceling microphone Monaural style Ensure call privacy with 64-bit digital encryption Extended hands-free talk time up to 10 full hours Roam up to 300 feet from your desk phone Over-the-head, comfortable design for all-day use Premium audio ensures quality customer communications Interactive Quick Start Guide
Svenska kronor2480.00

Plantronics CS70 headset
Modèle PCS70
Plantronics
The extremely stylish CS70 lets you use talk on the phone whilst up to 50m from your desk, but is so discreet and comfortable you may forget you are wearing it. With a clear voice tube microphone, it weighs just 22g but provides 5 hours talk-time. With the HL10 lifter or an electronic hook switch module, you can answer incoming calls when away from your telephone, helping to avoid calls unnecessarily going to voicemail. DECT audio technology ensures excellent audio quality and is compatible with most popular business telephones. Product Features: Incredibly light (22g) and comfortable DECT technology for excellent sound quality Compatible with most popular business telephones Volume and mute functions Boom swivels for use on either ear Up to 5 hrs talk time, 28hrs standby time with fast recharge. 100m range ? up to 50m in a typical office Unique removable voice tube for optimal hygiene Remote call answering when away from your telephone* IntelliStand? automatically answers call when headset is picked up from base* *- requires HL10 or electronic hookswitch control. Interactive Quick Start Guide
Svenska kronor2400.00

Sennheiser SH-320 headset
Modèle SH320
Sennheiser
When you have to concentrate and attention to detail is important, this two-in-one solution: monaural headset with temple pad or clip-headset helps you stay on top of the job. Light and comfortable to wear, you are free to work without distraction. Features reliable, high-quality performance and design that provides you with professional assistance in your office. With the adjustable and pivotable microphone boom, it is convenient to use and easy to transport. Active anti-shock in ear pieces safeguards against acoustic shock and hearing impairment with prolonged use. Features: Compatible with all major telephone systems Active anti-shock in ear pieces safeguards against acoustic shock and hearing impairment with prolonged use Headset can be worn on either left or right-hand side Comfortable, adjustable headband High-quality workmanship Frequency response curve optimized for excellent speech clarity Replaceable ear pad Sennheiser-specific easy disconnect facility
Svenska kronor1200.00

Sennheiser SH-350 headset
Modèle SH350
Sennheiser
SH350 Over the Head, Binaural Office Headset is light and comfortable to wear, you are free to work without distraction. Features reliable, high-quality performance and design that provides you with professional assistance in your office. With the adjustable and pivotable microphone boom, it is convenient to use and easy to transport. Active anti-shock in ear pieces safeguards against acoustic shock and hearing impairment with prolonged use. Features: Compatible with all major telephone systems Active anti-shock in ear pieces safegaurd against acoustic shock and hearing impairment with prolonged use Comfortable, adjustable headband Noise canceling microphone High-quality workmanship Modern, lightweight design for extended use Frequency response curve optimized for excellent speech clarity Sennheiser-specific easy disconnect facility
Svenska kronor1200.00

Snom HS-MM2 headset
Modèle HSMM2
Snom
for Snom 320, 360, 370 phones Flexible metal mic boom Monaural headset   Quick-disconnect cord RJ-11 modular plug Adjustable headband Noise-canceling microphone boom Soft leatherette ear cushion Superb audio quality Lightweight: 95g Headset hanger
Svenska kronor600.00

Snom HS-MM3 headset
Modèle HSMM3
Snom
Snom Headsets have been updated with a new look and improved ergonomics and usability. With the Snom headsets, snom provides a complete product line for use in a variety of Internet telephony environments: SMEs (small and medium-sized enterprises), home offices, and call centers. The Snom HS-MM3 headset fits perfectly to the snom 300 phone whereas the HS-MM2 headset is used for the Snom 320 and snom 360 phones. The new Snom HS-MM2 and HS-MM3 monaural headset provide users with more headset stability, ease of use, comfort, and hands-free convenience in their daily communication. The snom monaural headset has one ear pad which can be worn on either ear. With the noise- canceling microphone, boom the snom headset keeps your voice crystal clear.   The Snom Headsets are a safe investment for your business, providing users working in telephone intensive applications with more flexibility, freedom, and comfort. The use of headsets optimizes the operating process and increases work productivity.
Svenska kronor640.00

Voip accessories

Linksys PA-100EU power supply
Modèle PA-100
Linksys
The Linksys PA-100 Switching Power Supply provides A/C power to linksys SPA-9xx IP phones. This item is commonly purchased with the Linksys SPA942 or SPA922 IP phone because Linksys does not bundle power supplies with these phones.Detailed Compatible Model Numbers:Analog Telephone Adapters:Linksys SPA1001 Analog Telephone AdapterLinksys SPA2002 Analog Telephone AdapterLinksys SPA2100 Analog Telephone AdapterLinksys SPA2102 Analog Telephone AdapterLinksys SPA3000 Analog Telephone AdapterLinksys SPA3102 Analog Telephone AdapterLinksys PAP2 Analog Telephone AdapterLinksys PAP2T Analog Telephone AdapterLinksys WRTP54G Wireless Router + Analog Telephone AdapterIP Phones:Linksys SPA901 IP PhoneLinksys SPA921 IP PhoneLinksys SPA922 IP PhoneLinksys SPA941 IP PhoneLinksys SPA942 IP PhoneLinksys SPA962 IP Phone*The PA100 is also compatible with the "-NA" (unlocked) version of these model numbers.
Svenska kronor112.00

Product name	Category	Page
- Saturday 26 August,	Voip Gateways	54
Aastra 480i IP telephone - Friday 25 August, 20	IP Telephones	5
Aastra 9112i IP telefon - Friday 25 August, 20	IP Telephones	6
Allnet 7960 IP telephone - Saturday 05 May, 200	IP Telephones	6
Draytek 3300V router - Friday 04 August, 20	Voip Router	45
Elmeg IP290 IP Telephone - Friday 12 May, 2006	IP Telephones	7
Fiberlogic MeshVPN 2320 - Friday 27 October, 2	MeshVPN	60
Fiberlogic MeshVPN 2340 - Friday 27 October, 2	MeshVPN	61
Fiberlogic MeshVPN 2380 - Friday 27 October, 2	MeshVPN	63
GN NETCOM 9120 MidiBoom GAP headset - Friday 06 July, 2007	Headsets	71
GN Netcom 9330 headset - Friday 06 July, 2007	Headsets	72
Grandstream GXW-4008 FXS Analogue Gateway - Monday 12 February,	Voip Gateways	54
Grandstream GXW-4104 FXS Analogue Gateway - Monday 12 February,	Voip Gateways	55
Grandstream Budgetone 200 IP telephone - Sunday 18 March, 200	IP Telephones	7
Grandstream GXW-4108 FXS Analogue Gateway - Monday 12 February,	Voip Gateways	55
Grandstream GXW-4004 FXS Analogue Gateway - Monday 12 February,	Voip Gateways	54
Grandstream GXP 2000 IP Telephone - Friday 25 August, 20	IP Telephones	8
Grandstream GXP-2010 IP telephone - Sunday 28 October, 2	IP Telephones	9
Grandstream GXV-3000 - Friday 04 August, 20	IP Telephones	10
Granstream GXP-2020 IP telephone - Wednesday 02 May, 20	IP Telephones	11
Intertex SurfinBird IX67 FW GW Router - Thursday 11 May, 200	Voip Router	46
Intertex SurfinBird IX67 FW AIR GW Router - Friday 12 May, 2006	Voip Router	46
JUNGHANNS.NET quadBRI 2.0 ISDN card - Thursday 03 May, 200	ISDN cards	70
Lancom 1722 Voip - Friday 04 August, 20	Voip Gateways	56
Lancom 1724 Voip - Friday 04 August, 20	Voip Gateways	57
Level One VOI-7000 IP telephone - Saturday 05 May, 200	IP Telephones	12
Linksys CIT300 Skype teöephone - Saturday 18 November	Skype Telephones	37

Linksys WIP330 Wi-Fi telephone - Sunday 19 March, 200 Wi-Fi Telephones 30
Linksys PAP2T analogue adapter - Monday 08 January, 2Analogue Telephone 41
Linksys WRTP54G VoIP Wireless Router - Wednesday 19 April, Voip Router 53
Linksys SPA-922 IP Telephone - Wednesday 19 April, IP Telephones 14
Linksys WRT54GC router - Wednesday 16 August, Voip Router 52
Linksys RTP300 Broadband Router with 2 Phone Ports - Saturday 21 October, Voip Router 49
Linksys Wireless-G Broadband Router WRT54G - Wednesday 22 Februar Voip Router 50
Linksys Wireless-G Broadband Router WRT54GP2 - Saturday 11 February Voip Router 51
Linksys SPA-941 IP telephone - Friday 13 January, 2 IP Telephones 15
Linksys SPA-962 IP telephone - Wednesday 17 January IP Telephones 17
Linksys PA-100EU power supply - Monday 27 August, 20 Voip accessories 77
Linksys RT31P2 Voip Router - Thursday 11 May, 200 Voip Router 48
Linksys SPA-921 IP Telephone - Wednesday 19 April, IP Telephones 13
Linksys 24-port 10/100 + 2-Port GB Switch with POE - Saturday 12 August, Voip Router 47
Linksys SPA400 analogue adapter - Sunday 31 December, Analogue Telephone 44
Linksys SPA-9000 IP-PBX - Monday 20 March, 200 IP-PBX 58
Linksys CIT200 Skype telephone - Monday 20 March, 200 Skype Telephones 36
Linksys SPA-2102 analog - Saturday 14 January, Analogue Telephone 42
Linksys SPA3102 analogue adapter - Sunday 29 October, 2Analogue Telephone 43
Linksys SPA-942 IP telephone - Monday 20 March, 200 IP Telephones 16
LIP TA100 wi-fi telephone - Saturday 22 December Wi-Fi Telephones 31
Pirelli Discus DP-L10, Dual-mode Wi-Fi / GSM telephone - Saturday 24 FebruarySmartphones with V 40
Plantronics CS351 headset - Sunday 28 October, 2 Headsets 73
Plantronics CS70 headset - Sunday 28 October, 2 Headsets 74
Polycom SoundStation IP4000 - Friday 28 July, 2006 IP Telephones 21
Polycom SoundPoint IP 501 - Friday 28 July, 2006 IP Telephones 19
Polycom SoundPoint IP 601 - Friday 28 July, 2006 IP Telephones 20
Polycom SoundPoint IP 430 - Friday 28 July, 2006 IP Telephones 19
Polycom SoundPoint IP 320 IP telephone - Sunday 28 October, 2 IP Telephones 18
Polycom Communicator for Skype - Friday 16 March, 200 Skype Telephones 38
Polycom SoundPoint IP 301 - Friday 28 July, 2006 IP Telephones 17
Polycom SoundStation IP Extension Module - Friday 28 July, 2006 IP

Telephones 21
Sagem PhoneFax 49A - Tuesday 03 July, 200 Voip Fax 35
Sagem PhoneFax 43A - Wednesday 04 July, 2 Voip Fax 35
Sennheiser SH-350 headset - Sunday 28 October, 2 Headsets 75
Sennheiser SH-320 headset - Sunday 28 October, 2 Headsets 75
Siemens C450 IP cordless phone - Tuesday 12 December, DECT with VoIP 65
Siemens S450IP with GN Netcom 9120 headset - Tuesday 10 July, 200Companies Specials 2
Siemens OpenStage 60 IP telephone - Sunday 07 October, 2 IP Telephones 22
Siemens C470IP DECT telephone - Friday 04 January, 2 DECT with VoIP 67
Siemens OpenStage 40 IP telephone - Sunday 07 October, 2 IP Telephones 22
Siemens OpenStage 20 IP telephone - Sunday 07 October, 2 IP Telephones 22
Siemens OpenStage 80 IP telephone - Sunday 07 October, 2 IP Telephones 22
Siemens S450 IP cordless phone - Tuesday 12 December, DECT with VoIP 67
Siemens Gigaset S45 DECT telephone - Tuesday 27 March, 20 DECT with VoIP 67
Siemens 675 IP telephone - Saturday 22 Septembe DECT with VoIP 64
Siemens Gigaset SL75 WLAN - Sunday 19 November, Wi-Fi Telephones 32
Siemens Enterprise optiPoint WL2 professional S WLAN-Te - Friday 04 January, 2 Wi-Fi Telephones 32
Snom 300 IP Telephone - Sunday 30 July, 2006 IP Telephones 22
Snom 320 SIP Telephone - Thursday 12 January, IP Telephones 24
Snom 320 IP telephone & Headset HS-MM2 - Sunday 29 October, 2 IP Telephones 23
SNOM 360 Package Netcom 9350 - Tuesday 12 June, 200 Companies Specials 4
Snom 360 IP Telephone & Expansion Module - Wednesday 20 Septemb IP Telephones 26
SNOM 360 Package 360GN - Saturday 09 June, 20 Companies Specials 2
Snom 360 IP telephone - Thursday 12 January, IP Telephones 25
SNOM 370 Package 370GN - Thursday 07 June, 20 Companies Specials 4
Snom 370 IP telephone - Wednesday 02 May, 20 IP Telephones 27
SNOM 370 Package Midiboom - Tuesday 12 June, 200 Companies Specials 4
Snom HS-MM3 headset - Sunday 28 October, 2 Headsets 76
Snom HS-MM2 headset - Sunday 28 October, 2 Headsets 76
Snom M3 DECT IP telephone - Saturday 27 October, DECT with VoIP 68
TerraTec Noxon iRadio - Tuesday 13 March, 20 WiFi Radio 69
Thomson Speed Touch 2020 IP Telephone - Thursday 27 April, 2 IP Telephones 28
Thomson Speed Touch 2030 IP Telephone - Saturday 26 August, IP Telephones 28

Thomson SpeedTouch 2022 IP telephone - Saturday 09 June, 20 IP Telephones 29
UTStarcom F3000 Wi-Fi telephone - Sunday 28 May, 2006 Wi-Fi Telephones 33